



Yealink Gigabit Color IP Phone SIP-T46G User Guide

Version 81.20 Nov. 2016

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Statements of compliance can be obtained by contacting support@yealink.com.

CE Mark Warning

CE

This device is marked with the CE mark in compliance with EC Directives 2014/35/EU and 2014/30/EU.

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Any changes or modifications not expressly approved by the party responsible for compliance could void the user's authority to operate the equipment.

This device complies with Part 15 of the FCC Rules. Operation is subject to the following two conditions:

1. This device may not cause harmful interference, and

2. this device must accept any interference received, including interference that may cause undesired operation.

Industry Canada (IC)

This Class [B] digital apparatus complies with Canadian ICES-003 Rules.

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Note: This equipment has been tested and found to comply with the limits for a Class B digital device, pursuant to part 15 of the FCC Rules. These limits are designed to provide reasonable protection against harmful interference in a residential installation. This equipment generates, uses and can radiate radio frequency energy and, if not installed and used in accordance with the instructions, may cause harmful interference to radio communications. However, there is no guarantee that interference will not occur in a particular installation. If this equipment does cause harmful interference to radio or television reception, which can be determined by turning the equipment off and on, the user is encouraged to try to correct the interference by one or more of the following measures:

1. Reorient or relocate the receiving antenna.

- 2. Increase the separation between the equipment and receiver.
- 3. Connect the equipment into an outlet on a circuit different from that to which the receiver is connected.
- 4. Consult the dealer or an experienced radio/TV technician for help.

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To avoid the potential effects on the environment and human health as a result of the presence of hazardous substances in electrical and electronic equipment, end users of electrical and electronic equipment should understand the meaning of the crossed-out wheeled bin symbol. Do not dispose of WEEE as unsorted municipal waste and have to collect such WEEE separately.

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The original GPL license, source code of components licensed under GPL and used in Yealink products can be downloaded online:

http://www.yealink.com/GPLOpenSource.aspx?BaseInfoCateId=293&NewsCateId=293&CateId=293.

About This Guide

Thank you for choosing the SIP-T46G IP phone, an ultra-elegant Gigabit IP phone which is exquisitely designed to provide business telephony features, such as Call Hold, Call Transfer, Busy Lamp Field, Multicast Paging and Conference over an IP network.

This guide provides everything you need to quickly use your new phone. First, verify with your system administrator that the IP network is ready for phone configuration. Also be sure to read the Packaging Contents and Regulatory Notices sections in this guide before you set up and use the SIP-T46G IP phone.

Note

Network Directory and Network Call Log features are hidden for IP phones in neutral firmware version, which are designed for the BroadWorks environment. Please contact your system administrator for more information.

In This Guide

Topics provided in this guide include:

- Chapter 1 Overview
- Chapter 2 Getting Started
- Chapter 3 Customizing Your Phone
- Chapter 4 Basic Call Features
- Chapter 5 Advanced Phone Features

Summary of Changes

This section describes the changes to this guide for each release and guide version.

Changes for Release 81, Guide Version 81.20

Major updates have occurred to the following sections:

- Wallpaper on page 39
- Screen Saver on page 41

Changes for Release 81, Guide Version 81.15

The contents in the following section are removed to Frequently Asked Questions on Yealink IP

Phones.

Troubleshooting

The following section is new:

• BLF Ring Type on page 60

Changes for Release 81, Guide Version 81.5

The following sections are new:

- Network Connection on page 23
- Wireless Network Settings on page 28
- Screen Saver on page 41
- Power Saving on page 44
- Key Tone on page 59

Major updates have occurred to the following sections:

- Icon Instructions on page 3
- Basic Network Settings on page 26
- Volume on page 56
- Language on page 48
- DSS Keys on page 90
- Dial Plan on page 107
- Call Mute on page 127
- Do Not Disturb (DND) on page 130
- Call Transfer on page 143
- Call Park on page 150
- Busy Lamp Field (BLF) on page 163
- Call Recording on page 171
- Intercom on page 181
- Multicast Paging on page 185

Changes for Release 80, Guide Version 80.95

The following section is new:

• Entering Data and Editing Fields on page 36

Major updates have occurred to the following sections:

- Icon Instructions on page 3
- Optional Accessories on page 18
- Phone Installation on page 20

Changes for Release 80, Guide Version 80.80

Major updates have occurred to the following section:

• Bluetooth on page 85

Changes for Release 80, Guide Version 80.60

The following sections are new:

- Shared Call Appearance (SCA) on page 194
- Bridged Lines Appearance (BLA) on page 207

Major updates have occurred to the following sections:

- Icon Instructions on page 3
- LED Instructions on page 9
- Web User Interface on page 12
- Call Forward on page 135

Changes for Release 80, Guide Version 80.20

The following section is new:

• Short Message Service (SMS) on page 213

Major updates have occurred to the following sections:

- Icon Instructions on page 3
- Time & Date on page 49
- Bluetooth on page 85
- Anonymous Call Rejection on page 160
- Intercom on page 181
- Multicast Paging on page 185
- Appendix on page 223

Changes for Release 80, Guide Version 80.10

The following section is new:

• Optional Accessories on page 18

Major updates have occurred to the following sections:

- Phone Lock on page 53
- Local Directory on page 62
- Anonymous Call on page 159
- BLF List on page 169
- Appendix on page 223

Changes for Release 73, Guide Version 73.16

The following section is new:

• BLF List on page 169

Major updates have occurred to the following sections:

- Icon Instructions on page 3
- Ring Tones on page 57
- Anonymous Call Rejection on page 160
- Multicast Paging on page 185
- Appendix on page 223

Changes for Release 72, Guide Version 72.2

Major updates have occurred to the following sections:

- Documentations on page 15
- Packaging Contents on page 17

Changes for Release 72, Guide Version 72.1

Major updates have occurred to the following sections:

- Icon Instructions on page 3
- Backlight on page 47
- Contrast on page 48
- Anonymous Call on page 159

Changes for Release 71, Guide Version 71.180

Major updates have occurred to the following sections:

- Line Keys on page 90
- Hot Desking on page 180
- Appendix on page 223

Changes for Release 71, Guide Version 71.150

Major updates have occurred to the following sections:

- Bluetooth on page 85
- Remote Phone Book on page 77
- Line Keys on page 90
- Anonymous Call on page 159
- Appendix on page 223

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Overview

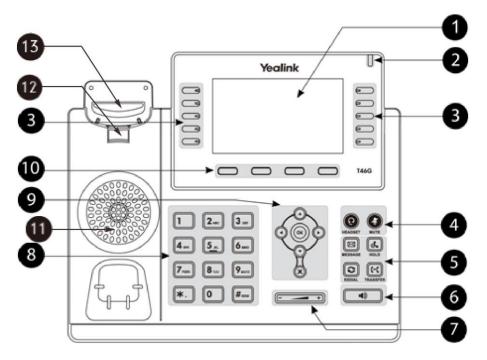
This chapter provides the overview of the SIP-T46G IP phone. Topics include:

- Hardware Component Instructions
- Icon Instructions
- LED Instructions
- User Interfaces
- Documentations

If you require additional information or assistance with your new phone, contact your system administrator.

Hardware Component Instructions

The main hardware components of the SIP-T46G IP phone are the LCD screen and the keypad.



	Item	Description
		Shows information about calls, messages, soft keys, time, date
1		and other relevant data:
		Default account
	LCD Screen	• Call information–caller ID, call duration
		• Icons (for example, 😑)
		Missed call text or second incoming caller information
		 Prompt text (for example, "Saving config file!") Time and date
		Indicates phone power and some feature statuses.
2	Power Indicator LED	Receives an incoming call—Fast flashing
(J)		Receives a voice mail or text message–Slowly flashing
		Use these keys to activate up to sixteen accounts and assign
3	Line Keys	various features.
	HEADSET Key	Toggles and indicates the headset mode.
4	MUTE Key	Toggles and indicates mute feature.
	MESSAGE Key	Accesses voice messages.
(5)	HOLD Key	Places a call on hold or resumes a held call.
9	REDIAL Key	Redials a previously dialed number.
	TRANSFER Key	Transfers a call to another party.
6	Speakerphone Key	Toggles and indicates the hands-free speakerphone mode.
7	Volume Key	Adjusts the volume of the handset, headset, speaker, ringer or media.
0	Kaunaal	Provides the digits, letters and special characters in
8	Keypad	context-sensitive applications.
	$\odot \odot \odot \odot$	Scroll through the displayed information.
9	ОК	Confirms actions or answers incoming calls.
	\mathbf{x}	Cancels actions or rejects incoming calls.
(10)	Soft Keys	Label automatically to identify their context-sensitive features.
(11)	Speaker	Provides hands-free (speakerphone) audio output.
		Secures the handset in the handset cradle when the IP phone
(12)	Hookswitch Tab	is mounted vertically. For more information on how to adjust
		the hookswitch tab, refer to Yealink Wall Mount Quick
		Installation Guide for Yealink IP Phones.
(13)	Hookswitch	Picking up the handset from the handset cradle, the
	hookswitch bounces and the phone connects to the line,	

Hardware component instructions of the SIP-T46G IP phone are:

Item	Description
	laying the handset down on the handset cradle, the phone
	disconnects from the line.

Icon Instructions

Feature Status Icons

Icons appearing on the status bar are described in the following table:

Icons	Description
	Network is unavailable
40	Hands-free speakerphone mode
C	Handset mode
Ç	Headset mode
	Voice Mail
	Text Message
A _A	Auto Answer
•	Do Not Disturb (DND)
\$	Keep Mute
*	Ringer volume is 0
	Phone Lock
-	Missed Calls
5	Call Forward
۲	Recording starts successfully (Using a USB flash drive)
U	Recording is paused (Using a USB flash drive)

Icons	Description
*	Bluetooth mode is on
*	Bluetooth headset is both paired and connected
0)	Wi-Fi connection is successful
×	Wi-Fi connection failed

Line Key Icons

Icon indicator (associated with line key features)

Icons	Description
	Hold
	Private Hold
•	DND
@@	Voice Mail
	SMS
®	Direct Pickup
8	Group Pickup
K iii	DTMF
`	Prefix
北	Local Group
	XML Group
1	LDAP
3	XML Browser
2	Conference
5	Forward
**	Transfer
8	ReCall

Icons	Description
	Record
•	URL Record
۲	Recording starts successfully (Record/URL Record)
	Multicast Paging
2	Group Listening
	Paging List
8	Hot Desking
Ŷ	Zero Touch
	URL
	Phone Lock
2	Directory
000	Speed Dial

Icon indicator (associated with line)

Icons	Description
6	The private line registers successfully
P	The shared/bridged line registers successfully
1	Registering
(ja	Register failed
2	DND is enabled on this line
6	Call forward is enabled on this line

Icon indicator (associated with call park/retrieve park)

Icons	Description
C.	Call Park (Park successfully; Call park idle state)
92	Call Park (Park failed)

Icons	Description	
¢.	Call Park (Call park ringing state)	
ð	Retrieve Park	

Icon indicator (associated with ACD)

Icon	Description
Ð	Logged in the ACD system
3	The ACD status is available
4	The ACD state is Wrap up
28	The ACD status is unavailable
G	Logged out of the ACD system

Icon indicator (associated with intercom)

Icons	Description
a	Intercom idle state
	Intercom ringing state
	Intercom callout/talking state
	Intercom failed state

Icon indicator (associated with BLF/BLF List)

Icons	Description
2	BLF/BLF List idle state
(Flashing)	BLF/BLF list ringing state
(Flashing)	BLF/BLF list callout state

Icons	Description
2	BLF/BLF list talking state
2	BLF hold state
20	BLF list call park state
2	BLF/BLF List failed state

Icon indicator (associated with a shared line)

The local SCA phone indicates the phone is involved in an SCA call, while the monitoring SCA phone indicates the phone is not involved in the SCA call.

Icon	Description
2	The shared line is idle.
(for monitoring SCA phone)	The shared line is seized.
2	The shared line receives an incoming call.
2	The shared line is dialing.
	The shared line is in conversation.
20	The shared line conversation is placed on public hold.
(for local SCA phone)	The shared line conversation is placed on private hold.
<u></u>	The shared line conversation is barged in by the other shared line party.
	For a multi-party call, all the shared line parties participating in this call place the shared line conversation on hold.

Icon indicator (associated with a bridged line)

The local BLA phone indicates the phone is involved in a BLA call, while the monitoring BLA phone indicates the phone is not involved in the BLA call.

Icon	Description
*	The bridged line is idle.
% (for monitoring BLA phone)	The bridged line is seized.
2	The bridged line receives an incoming call.
1	The bridged line is dialing.
2	The bridged line is in conversation.
2	The bridged line conversation is placed on hold.

Other Icons

Icon indicator (associated with call recording)

Icon	Description
\$	Recording box is full (Using a record key)
R	A call cannot be recorded (Using a record key)
۲	Recording starts successfully (Using a record key)
X	Recording cannot be started (Using a record key)
X	Recording cannot be stopped (Using a record key)

Icon indicator (associated with call history)

Icon	Description
+	Received Calls

Icon	Description	
Ť	Placed Calls	
	Missed Calls	
5	Forwarded Calls	

Icon indicator (associated with contact)

Icon	Description	
<u></u>	Contact icon	
	The default caller photo	

LED Instructions

Power Indicator LED

LED Status	Description	
Solid red	The phone is initializing.	
Fast flashing red (300ms)	The phone is ringing.	
Slowly flashing red (1s)	The phone receives a voice mail or text message.	
	The phone is powered off. The phone is idle.	
Off	The phone is busy.	
	The call is placed on hold or is held.	
	The call is muted.	

Line key LED

LED Status	Description	
Solid groop	The line is seized.	
Solid green	The line is in conversation.	
Fast flashing green	The line receives an incoming call.	
Slowly flashing green	The call is placed on hold.	
Off	The line is inactive.	

Line key LED (configured as a BLF key or BLF List key)

LED Status	Description	
Solid green	The monitored user is idle.	
Fast flashing red (200ms)	The monitored user receives an incoming call.	
Solid red	The monitored user is busy. The monitored user's conversation is placed on hold (This LED status requires server support).	
Slowly flashing red (1s)	The call is parked against the monitored user's phone number.	
Off	The monitored user does not exist.	

Line key LED (used as a page switch key)

LED Status	Description	
Solid red	There is a parked call on the line key of the non-current page.	
Fast flashing green	The line receives an incoming call on the line key of the non-current page.	
Fast flashing red	The monitored user receives an incoming call on the line key of the non-current page.	
Off	The line keys are idle.	

Line key LED (associated with a shared line)

The local SCA phone indicates the phone is involved in an SCA call, while the monitoring SCA phone indicates the phone is not involved in the SCA call.

LED Status	Description	
Off	The shared line is idle.	
Fast flashing green	The shared line receives an incoming call.	
Solid green (for local SCA phone) Solid red (for monitoring SCA phone)	The shared line is in conversation. The shared line is dialing. The shared line is seized. The shared line conversation is barged in by the other shared line party.	
Slowly flashing green (for local SCA phone) Slowly flashing red (for monitoring SCA phone)	The shared line conversation is placed on public hold.	
Slowly flashing green (for local SCA phone) Solid red (for monitoring SCA phone)	The shared line conversation is placed on private hold. For a multi-party call, all the shared line parties participating in this call place the shared line conversation on hold.	

Line key LED (associated with a bridged line)

The local BLA phone indicates the phone is involved in a BLA call, while the monitoring BLA phone indicates the phone is not involved in the BLA call.

LED Status	Description	
Off	The bridged line is idle.	
Fast flashing green	The bridged line receives an incoming call.	
Solid green (for local BLA phone) Solid red (for monitoring BLA phone)	The bridged line is in conversation. The bridged line is dialing. The bridged line is seized.	
Slowly flashing green (for local BLA phone) Slowly flashing red (for monitoring BLA phone)	The bridged line conversation is placed on hold.	

Note

The above introduces the default LED status. The statuses of the power indicator LED and BLF key LED are configurable via web user interface. You can also customize the BLF key LED status and BLF key behavior. For more information, contact your system administrator.

User Interfaces

Two ways to customize configurations of your SIP-T46G IP phone:

- The user interface on the IP phone.
- The user interface in a web browser on your PC.

The hardware components keypad and LCD screen constitute the phone user interface, which allows the user to execute all call operation tasks and basic configuration changes directly on the phone. In addition, you can use the web user interface to access all configuration settings. In many cases, it is possible to use either the phone user interface and/or the web user interface interchangeably. However, in some cases, it is only possible to use one or the other interface to operate the phone and change settings.

Phone User Interface

You can customize your phone by pressing the **Menu** soft key to access the phone user interface. The Advanced option is only accessible to the administrator, and the default administrator password is "admin" (case-sensitive). For more information on customizing your phone with the available options from the phone user interface, refer to Customizing Your Phone on page 39.

Note For a better understanding of the menu structure, please refer to Appendix B – Menu Structure on page 225.

Web User Interface

In addition to the phone user interface, you can also customize your phone via web user interface. In order to access the web user interface, you need to know the IP address of your new phone. To obtain the IP address, press the OK key on the phone when the phone is idle. Enter the IP address (e.g., http://192.168.0.10 or 192.168.0.10 for IPv4; http://[2005:1:1:1:215:65ff:fe64:6e0a] or [2005:1:1:1:215:65ff:fe64:6e0a] for IPv6) in the address bar of a web browser on your PC. The default administrator user name and password are both "admin" (case-sensitive).

Note

The access of Advanced settings page of the Account or Network web page may be restricted by the web explorer (e.g., Chrome, Firebox) if you have installed "Adblock Plus plugin".

Options	Phone User Interface	Web User Interface
Status		
IPv4		
MAC		
Firmware	\checkmark	\checkmark
Network		
Phone		
Accounts		
Basic Phone Settings		
Wallpaper	\checkmark	
Screen Saver	\checkmark	\checkmark
Power Saving	Х	
Backlight	\checkmark	
Contrast	\checkmark	×
Language	\checkmark	/

The options you can use to customize the IP phone via phone user interface and/or via web user interface are listed in the following table:

Options	Phone User Interface	Web User Interface
Time & Date	~	
Administrator Password	\checkmark	
Key As Send	\checkmark	
Phone Lock	\checkmark	
Audio Settings		
Ring Tones	\checkmark	
Key Tone	\checkmark	
BLF Ring Type	\checkmark	
Contact Management		
Directory	х	
Local Directory	\checkmark	
Blacklist	\checkmark	
Remote Phone Book	x	
Call History Management	~	
Search Source List in Dialing	х	
Headset Use	\checkmark	
Bluetooth Headset	\checkmark	
DSS Keys	\checkmark	
Account Registration	\checkmark	
Dial Plan	х	
Emergency Number	х	
Live Dialpad	х	
Hotline	~	
Basic Call Features		
Recent Call In Dialing	х	
Auto Answer	\checkmark	
Auto Redial	\checkmark	
Call Completion	\checkmark	
ReCall	\checkmark	
Do Not Disturb (DND)	~	
Call Forward	\checkmark	\checkmark
Call Transfer	\checkmark	
Call Waiting	~]
Conference	х	
Call Park	\checkmark	
Call Pickup	~]
Anonymous Call	~	
Anonymous Call Rejection	~	
Advanced Phone Features		
Busy Lamp Field (BLF)	√	√
BLF List	x	

Options	Phone User Interface	Web User Interface
Call Recording	~	
Hot Desking	\checkmark	
Intercom	\checkmark	
Multicast Paging	x	
Music on Hold	x	
Automatic Call Distribution (ACD)	х	
Shared Call Appearance (SCA)	x	
Bridged Lines Appearance (BLA)	x	
Messages	~	
SIP Account		
User Options		
Activation	\checkmark	
Label	~	
Display Name	~	
Register Name	~	
User Name	~	
Password	~	\checkmark
Server Options		
SIP Server 1/2	~	
Register Port	x	
Outbound Status	~	
Outbound Proxy1/2	~	
Fallback Interval	~	
NAT Status	~	

Note

The table above lists most of the feature options. Please refer to the relevant sections for more information.

Documentations

Name	Contents	Where found	Language
Quick Start Guide	Basic call features and phone	In the package	English
Quick start Guide	customizations	On the website	English/Chinese
User Guide	Phone/Web user interface settings, basic call features and advanced phone features	On the website	English/Chinese

The following table shows documentations available for the SIP-T46G IP phone.

Note

You can also download the latest documentations online: http://support.yealink.com/documentFront/forwardToDocumentDetailPage?documentId=10.

Getting Started

This chapter provides the following basic installation instructions and information for obtaining the best performance with the SIP-T46G IP phone. Topics include:

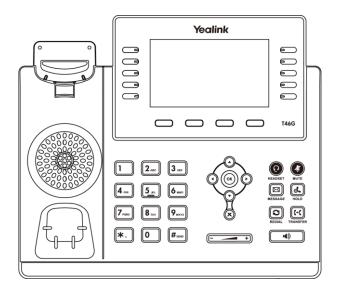
- Packaging Contents
- Phone Installation
- Network Connection
- Phone Initialization
- Phone Status
- Basic Network Settings
- Wireless Network Settings
- Registration
- Idle Screen
- Entering Data and Editing Fields

If you require additional information or assistance with your new phone, contact your system administrator.

Packaging Contents

The following components are included in your SIP-T46G IP phone package:

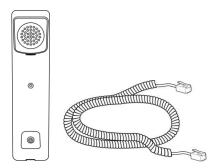
• SIP-T46G IP Phone



• Phone Stand



• Handset & Handset Cord



• Ethernet Cable (2m CAT5E UTP cable)



• Quick Start Guide

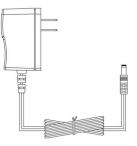


Check the list before installation. If you find anything missing, contact your system administrator.

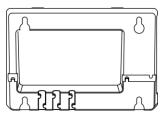
Optional Accessories

The following items are optional accessories for your SIP-T46G IP phone. You need to purchase them separately if required.

• Power Adapter (5.5mm DC plug)



• Wall Mount Bracket



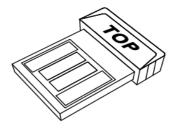
• Headset



• Wireless Headset Adapter EHS36



• Bluetooth USB Dongle BT40



• Wi-Fi USB Dongle WF40



Note

We recommend that you use the accessories provided or approved by Yealink. The use of unapproved third-party accessories may result in reduced performance.

Phone Installation

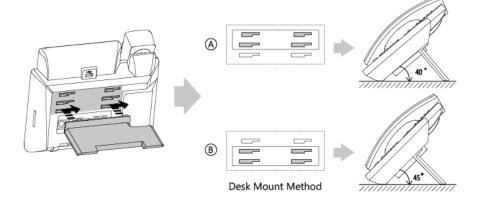
If your phone is already installed, proceed to Phone Installation on page 20.

This section introduces how to install the phone:

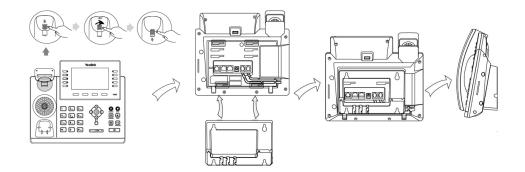
- 1) Attach the stand or the optional wall mount bracket
- 2) Connect the handset and optional headset
- 3) Connect the optional USB flash drive
- 4) Connect the power

1) Attach the stand or the optional wall mount bracket

Desk Mount Method



Wall Mount Method (Optional)

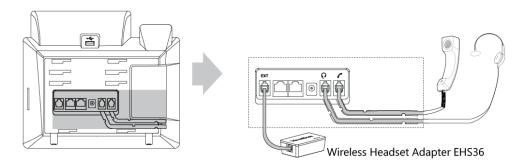


Note

The hookswitch tab has a lip which allows the handset to stay on-hook when the IP phone is mounted vertically.

For more information on how to attach the wall mount bracket, refer to *Yealink Wall Mount Quick Installation Guide for Yealink IP Phones.*

2) Connect the handset and optional headset

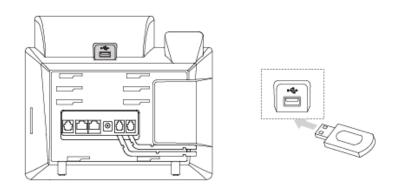


Note

The EXT port can also be used to connect an expansion module EXP40.

For more information on how to use EHS36 and EXP40, refer to *Yealink EHS36 User Guide* and *Yealink EXP40 User Guide*.

3) Connect the optional USB flash drive



Note

The USB flash drive should be purchased separately if required.

The USB port can also be used to connect the Bluetooth USB dongle BT40 or Wi-Fi USB dongle WF40. The Bluetooth USB dongle BT40 and Wi-Fi USB dongle WF40 should be purchased separately. For more information on how to use the BT40 and WF40, refer to *Yealink Bluetooth USB Dongle BT40 User Guide* and *Yealink Wi-Fi USB Dongle WF40 User Guide*.

4) Connect the power

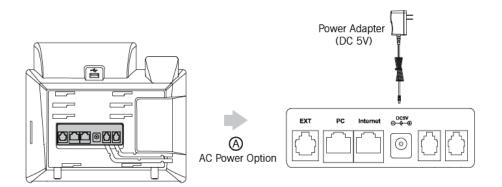
You have two options for power connections. Your system administrator will advise you which one to use.

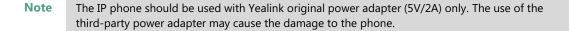
- AC power (Optional)
- Power over Ethernet (PoE)

AC Power (Optional)

To connect the AC power:

1. Connect the DC plug on the power adapter to the DC5V port on the phone and connect the other end of the power adapter into an electrical power outlet.



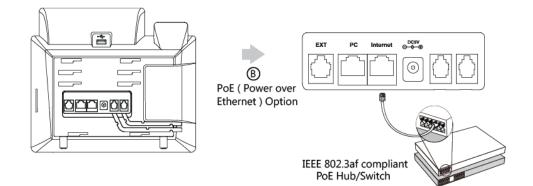


Power over Ethernet

With the included or a regular Ethernet cable, the SIP-T46G IP phone can be powered from a PoE-compliant switch or hub.

To connect the PoE:

1. Connect the Ethernet cable between the Internet port on the phone and an available port on the in-line power switch/hub.



Note If in-line power is provided, you don't need to connect the phone to the power adapter. Make sure the switch/hub is PoE-compliant.

Important! Do not remove power from the phone while it is updating firmware and configurations.

Network Connection

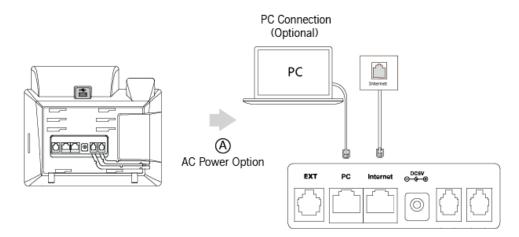
You have two options for network connection. You can select one of them according to the current office environment.

- Connecting to the wired network
- Connecting to the wireless network

Connecting to the Wired Network

You can connect your phone to a wired network.

Before connecting your phone to the wired network, it is important to note that the Wi-Fi feature should be disabled. For more information, refer to Connecting to the Wireless Network on page 24.



Note The phone can also share the network with another network device such as a PC (personal computer). This is an optional connection. For more information on this function, contact your system administrator.

Connecting to the Wireless Network

Yealink SIP-T46G IP phone supports Wi-Fi feature. You can connect your phone to a wireless network. The wireless network is more convenient and cost-effective than wired network. The PC port is unavailable when using wireless network. For more information on how to connect to a wireless network, refer to Wireless Network Settings on page 28.

Phone Initialization

After your phone is powered on, the system boots up and performs the following steps:

Automatic Phone Initialization

The phone finishes the initialization by loading the saved configuration. The LCD screen displays "Welcome Initializing...Please wait" during the initialization.

DHCP (Dynamic Host Configuration Protocol)

The phone attempts to contact a DHCP server in your network to obtain valid IPv4 network settings (e.g., IP address, subnet mask, default gateway address and DNS address) by default.

If your network does not use DHCP, proceed to Basic Network Settings on page 26.

Phone Status

Note

You can view phone status via phone user interface or web user interface.

Available information of phone status includes:

- Network status (IPv4 status or IPv6 status, IP mode and MAC address).
 - IPv4 uses a 32-bit address.
 - IPv6 is an updated version of the current Internet Protocol to meet the increased demands for unique IP addresses, using a 128-bit address.
- Phone status (product name, hardware version, firmware version, product ID, MAC address and device certificate status).
- Account status (register status of SIP accounts).

To view the phone status via phone user interface:

- **1.** Press (or), or press **Menu->Status**.
- **2.** Press (\bullet) or (\bullet) to scroll through the list and view the specific information.

	Status
1. IPv4:	10.2.20.210
2. MAC:	00-15-65-41-46-DD
3. Firmware:	28.81.0.20
4. More	

To view the phone status via web user interface:

- 1. Open a web browser on your computer.
- Enter the IP address in the browser's address bar, and then press the Enter key.
 (e.g., "http://192.168.0.10" for IPv4 or "http://[2005:1:1:1:215:65ff:fe64:6e0a]" for IPv6).

3. Enter the user name (admin) and password (admin) in the login page.

Login	Gigabit Color IP Phone SIP-T46G		
Username Password			
Cor	nfirm Cancel		

4. Click Confirm to login.

The phone status is displayed on the first page of the web user interface.

Yealink 1466	Status Account Network	Dsskey Features	Log Out Englah(Englah) - Settings Directory Security
Status	Version 🕜		NOTE
	Firmware Version	28.81.0.20	Version
	Hardware Version	28.0.0.128.0.0.0	It shows the version of firmware and hardware.
	Device Certificate 🛛 🥝		
	Device Certificate	Factory Installed	Network It shows the network settings
	Network 🕜		of Internet (WAN) port.
	Internet Port	IPv4	Account It shows the registration status
	IPv4 🕜		of SIP accounts.
	WAN Port Type	DHCP	You can click here to get more guides.
	WAN IP Address	10.2.20.210	
	Subnet Mask	255.255.255.0	
	Gateway	10.2.20.254	
	Primary DNS	192.168.1.20	
	Secondary DNS	192.168.1.22	

Note If IP mode of the phone is configured as **IPv4 & IPv6**, you can enter either of them in the browser's address bar to view the phone status. IPv6 is not available on all servers. Contact your system administrator for more information.

Basic Network Settings

If your phone cannot contact a DHCP server for any reason, you need to configure network settings manually. The IP phone can support either or both IPv4 and IPv6 addresses.

To configure the IP mode via phone user interface:

- 1. Press Menu->Advanced (default password: admin) ->Network->WAN Port.
- 2. Press (•) or (•), or the Switch soft key to select IPv4, IPv6 or IPv4 & IPv6 from the IP Mode field.

WAN Port				
1. IP Mode:	IPv4	$\triangleleft \triangleright$		
2. IPv4				
3. IPv6				
Back	Switch	Save		

3. Press the Save soft key to accept the change or the Back soft key to cancel.

You can configure a static IPv4 address for the IP phone. Before configuring it, make sure that the IP mode is configured as **IPv4** or **IPv4 & IPv6**.

To configure a static IPv4 address via phone user interface:

- 1. Press Menu->Advanced (default password: admin) ->Network->WAN Port.
- **2.** Press (\bullet) or (\bullet) to select **IPv4** and then press the **Enter** soft key.

	WAN Port	
1. IP Mode:	IPv4	$\triangleleft \triangleright$
2. IPv4		
3. IPv6		
Back		Enter

- **3.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select the **Static IP** from the **Type** field.
- Enter the desired value in the IP Address, Subnet Mask, Default Gateway, Pri.DNS and Sec.DNS field respectively.

IPv4 WAN Port				
1. Type:		Static IP	$\triangleleft \triangleright$	
2. IP Address:		192.168.1.10		
3. Subnet Mas	k:	255.255.255.0		
4. Default Gateway: 192.168.1.1				
5. Pri.DNS: 202.101.103.35				
Back	123	Delete	Save	

5. Press the Save soft key to accept the change or the Back soft key to cancel.

You can configure a static IPv6 address for the IP phone. Before configuring it, make sure that the IP mode is configured as **IPv6** or **IPv4 & IPv6**.

To configure a static IPv6 address via phone user interface:

- 1. Press Menu->Advanced (default password: admin) ->Network->WAN Port
- **2.** Press (\bullet) or (\bullet) to select **IPv6** and then press the **Enter** soft key.

	WAN Port	
1. IP Mode:	IPv6	$\triangleleft \triangleright$
2. IPv4		
3. IPv6		
Back		Enter

- 3. Press (\bullet) or (\bullet), or the Switch soft key to select the Static IP from the Type field.
- Enter the desired value in the IP Address, IPv6 IP Prefix, Default Gateway, Pri.DNS and Sec.DNS field respectively.

IPv6 WAN Port			
1. Type:		Static IP	$\triangleleft \triangleright$
2. IP Address	:	2005:1:1:1::25	
3. IPv6 IP Prefix:		64	
4. Default Gateway:		2005:1:1:1::25	
5. Pri.DNS: 2005:1:1:1::89			
Back	123	Delete	Save

5. Press the Save soft key to accept the change or the Back soft key to cancel.

The wrong network settings may result in inaccessibility of your phone and may also have an impact on your network performance. For more information on these parameters, contact your system administrator.

Wireless Network Settings

Wi-Fi is a popular wireless networking technology that uses radio waves to provide wireless high-speed Internet and network connections. You can activate/deactivate the Wi-Fi mode on the IP phone, and then connect your phone to the desired wireless network.

Note Ensure that the Wi-Fi USB Dongle WF40 is properly connected to the USB port at the back of the IP phone. For more information, refer to Phone Installation on page 20.

Note

Activating the Wi-Fi Mode

To activate the Wi-Fi mode via phone user interface:

- 1. Press Menu->Basic->Wi-Fi.
- **2.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select **On** from the **Wi-Fi** field.
- 3. Press the Save soft key.

The IP phone scans the available wireless networks in your area.

Wi-Fi				
Wi-Fi: On < 🕨				
Wireless Statu	s			
The storage n	etwork			
30 Available N	etwork(s)(Cor	nnected with TP	-LIN	
Back	WPS	Switch	Save	

When Wi-Fi USB dongle WF40 is connected to the USB port at the back of the IP phone, the LCD screen will prompt "Wi-Fi adapter has been added, scanning the available wireless network right now?". You can press the **OK** soft key to activate Wi-Fi mode directly and scan Wi-Fi devices automatically.



Connecting to the Wireless Network

Three ways to connect SIP-T46G IP phone to the wireless network:

- Manually connect to an available wireless network
- Wi-Fi Protected Setup (WPS)
- Manually add a wireless network

Manually Connect to an Available Wireless Network

To manually connect to a wireless network via phone user interface:

- 1. Press Menu->Basic->Wi-Fi.
- 2. Press or to highlight X Available Network(s) (X represents the number of available networks), and then press the Enter soft key.

If there is no available wireless network, press the **Scan** soft key to search for the wireless network again.

3. Press (\bullet) or (\bullet) to select the desired wireless network.

Available Wireless Network				
Yealink_Me	✓ 🔒 ≬			
and-Busine	🧇 and-Business			
🤝 Cert				
Sector CMCC				
CMCC-WEB				
Back	Detail	Scan	Connect	

- 4. Press the **Connect** soft key.
- 5. If the network is secure, enter its password in the **Password** field.

	Available Wir	eless Network	
Yealink_Me	eting		A
🐡 ar 🚺 PIN			
Pass CI CMCC-WEB	sword $\bullet \bullet \bullet $		
Cancel	123	Delete	ОК

6. Press the **OK** soft key.

Once the connection has completed successfully, the icon \checkmark appears after the name of the wireless network.

Available Wireless Network				
Yealink_Meeting			A	
🛜 and-Business				
🤝 Cert				
CMCC				
CMCC-WEB				
Back	Detail	Scan	Disconnect	

The Wi-Fi icon 🖙 appears on the status bar.



Note

For more information on the password of the wireless network, contact your system administrator.

Wi-Fi Protected Setup (WPS)

Wi-Fi Protected Setup (WPS) provides simplified mechanisms to configure secure wireless networks. WPS can automatically configure a wireless network with a network name (SSID) and strong WPA data encryption and authentication.

Two methods supported by Yealink IP phones in the Wi-Fi protected setup:

- Push Button Configuration (PBC): The user simply has to press WPS key on both the IP phone and gateway/router to connect.
- **Personal Identification Number (PIN)**: The user has to enter a PIN generated randomly by the IP phone on the gateway/router to connect.

Push Button Configuration (PBC)

To connect to the wireless network using PBC method:

- 1. Press Menu->Basic->Wi-Fi.
- Press the WPS soft key, and then select WPS from the prompt list.
 The LCD screen prompts "WPS has opened, please connect in 120s.".
- 3. Long press the WPS key on your gateway/router.

Once WPS setup has completed successfully, the icon \checkmark appears after the name of wireless network.

Personal Identification Number (PIN)

To connect to the wireless network using PIN method:

- 1. Press Menu->Basic->Wi-Fi.
- 2. Press the WPS soft key, and then select WPS-PIN from the prompt list.

The LCD screen prompts "Please input below PIN code on your Wi-Fi router and press OK to start connecting: xxx".

 Log into your gateway/router's web interface, and configure it to search the clients. For more information, refer to the documentation from the gateway/router manufacturer.
 Once WPS-PIN setup has completed successfully, the gateway/router's web interface will prompt success.

Note Contact your gateway/router provider for more information.

Manually Add a Wireless Network

If your gateway/router has SSID broadcast disabled, it might not appear in the scanning results. In that case, you must manually add a wireless network.

To add a wireless network:

- 1. Press Menu->Basic->Wi-Fi.
- **2.** Press (\bullet) or (\bullet) to highlight **The storage network**, and then press the **Add** soft key.
- **3.** Use the WLAN settings obtained from your gateway/router to configure this WLAN Profile on the IP phone. Do the following:
 - a) If you select **None** or **WEP** from the pull-down list of **Security Mode**:

Enter the profile name, SSID and WPA shared key in the corresponding fields.

b) If you select WPA-PSK or WPA2-PSK from the pull-down list of Security Mode:

Select the desired Cipher type (**TKIP**, **AES** or **TKIP AES**) from the pull-down list of **Cipher Type**.

Enter the profile name, SSID and WPA shared key in the corresponding fields.

4. Press the **Save** soft key to accept the change or the **Back** soft key to cancel.

The connected wireless networks will be saved to the storage network list. The storage network list can store up to 5 wireless networks, you can specify the priority to connect to the corresponding wireless network.

Viewing the Wireless Network Information

To view the connected wireless network information via phone user interface:

- Press Menu->Basic->Wi-Fi->X Available Network(s) (X represents the number of available networks).
- **2.** Press (\bullet) or (\bullet) to highlight the desired wireless network.
- **3.** Press the **Detail** soft key to view the detailed wireless network information (e.g., Profile Name, SSID or Signal Strength).

If the IP phone is connected to a wireless network successfully, you can also press **Menu->Basic->Wi-Fi->Wireless Status** to view the connected wireless network information.

Wireless Status			
1. Wireless Status:	Connected		
2. Profile Name:	didiaodewifi		
3. SSID:	didiaodewifi		
4. Signal Strength:	4		
5. Channel:	6		
Back			

Managing the Saved Wireless Network

Once the IP phone has ever been connected to a wireless network successfully, the wireless network profile will be saved in **The storage network** list. The next time you enable Wi-Fi feature, the IP phone will be automatically connected to the wireless network which has the highest priority (the top one). Up to 5 wireless network profiles can be saved in **The storage network** list. If you want to add a new one when the IP phone has already saved 5 wireless network profiles, please delete the older saved profile before adding.

To manage the storage network via phone user interface:

1. Press Menu->Basic->Wi-Fi->The storage network.

	Storage I	Network	
Yealink_Meet	ting		✓ 🔒
and-Business	;		
TP-LINK_1C	7C		
Back	Option	Add	Disconnect

- 2. Press the **Option** soft key.
- **3.** Do one of the following:
 - To edit the wireless network profile:
 - 1) Select Edit from the prompt list.
 - 2) Edit the desired wireless network information.
 - 3) Press the Save soft key.
 - To change the priority of the wireless network:

- 1) Select **Move Up** or **Move Down** from the prompt list to improve or lower the priority of the wireless network that the IP phone automatically connects to.
- To delete the saved wireless network profile(s):
 - 1) Select Delete or Delete All from the prompt list.

If you delete your wireless network profile from your phone, you need to manually connect to it next time.

Disconnecting the Wireless Network Connection

To disconnect wireless network via phone user interface:

- Press Menu->Basic->Wi-Fi->X Available Network(s) (X represents the number of available networks)
- **2.** Select the connected wireless network, and then press the **Disconnect** soft key.

The wireless network is disconnected. You can press the **Connect** soft key to connect again.

Deactivating the Wi-Fi Mode

To deactivate the Wi-Fi mode via phone user interface:

- 1. Press Menu->Basic->Wi-Fi.
- **2.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select **Off** from the **Wi-Fi** field.

	Wi-Fi	
Wi-Fi:	Off	$\langle \rangle$
Wireless Status		
Back	Switch	Save

3. Press the Save soft key to accept the change.

The Wi-Fi icon 🖙 disappears from the status bar.

Wi-Fi is configurable via web user interface at the path Network->Wi-Fi.

Registration

Generally, your phone will be deployed with multiple other phones. In this case, your system administrator will configure the phone parameters beforehand, so that after you start up your phone, the phone will be registered and ready for use. The SIP-T46G IP phone supports up to 16

accounts. If your phone is not registered, you may have to register it. For more information on how to register your phone, refer to Account Management on page 104.

Idle Screen



If the phone has successfully started up, the idle LCD screen will be displayed as below.

Idle	screen	description:

No.	Name	Description
1	Ctatus Day	This area shows the phone's default account, icons and the time and
1	Status Bar	date.
		The label of the default account is displayed on the left of the status
	Default	bar.
2	Account	For more information, refer to Default Account on page 106.
	Account	Note: You can configure the IP phone not to display the default
		account. Contact your system administrator for more information.
3	Icons	Icons are displayed in the middle of the status bar.
5	ICONS	For more information, refer to Icon Instructions on page 3.
	Time and	The phone's time and date are displayed on the right of the status
4	Date	bar.
	Date	For more information, refer to Time & Date on page 49.
		This area shows the line key labels. You can customize the line key as
5	Line Keys	other functions.
		For more information, refer to Line Keys on page 90.
6	Page Icons	This area shows three page icons of line keys.
0	i age icons	For more information, refer to Page Tips on page 92.
7	Soft Keys	This area shows the soft key labels. The default soft key labels are
/	SOIL REYS	"History", "Directory", "DND" and "Menu".
		This area shows the wallpaper which can be customized.
8	Wallpaper	For more information, refer to Wallpaper on page 39.
		Note: You can also enable the IP phone to display a large DND icon

No.	Name	Description
		on the idle screen. For more information, refer to Do Not Disturb
		(DND) on page 130

Entering Data and Editing Fields

You can enter data and edit fields using the phone keypad.

Keypad on the phone provides different characters (or numbers) when using the **2aB**, **abc**, **Abc**, **ABC** or **123** input mode. You can change the following input modes to enter data and edit fields on your phone. When your phone keypad matches the input mode, you can press the keypad repeatedly to view the character (or number) options and stop to select. When the character (or number) you want to enter displays in the field, wait one second, and enter the next character (or number).

Input Mode Keypad	2aB	abc	Abc (initials in capitals)	ABC	123
1	1				1
	2abcABC	abc2äæåàá âãç	abc2äæåàá âãç	ABC2ÄÆÅ ÀÁÂÃÇ	2
3 DEF	3defDEF	def3èéêëð	def3èéêëð	DEF3ÈÉÊËÐ	3
4 GHI	4ghiGHI	ghi4ìíîï	ghi4ìíîï	GHI4ÌÍÎÏ	4
5_JKL	5jklJKL	jkl5£	jkl5£	JKL5£	5
6 MINO	6mnoMNO	mno6öøòó ôõñ	mno6öøòó ôõñ	MNO6ÖØ ÒÓÔÕÑ	6
7 Pars	7pqrsPQRS	pqrs7ßS	pqrs7ßS	PQRS7S	7
8 тих	8tuvTUV	tuv8ùúûü	tuv8ùúûü	TUV8ÙÚÛ Ü	8
9 _{wxvz}	9wxyzWXY Z	wxyz9ýÞ	wxyz9ýÞ	WXYZ9ÝÞ	9
0	0	space	space	space	0

The following table lists the input modes and character (or number) options for the keypad:

Input Mode Keypad	2aB	abc	Abc (initials in capitals)	ABC	123
*.	*.,'?!\-()@/: _;+&%=<> £ \$¥¤[]{}~ ^j¿§#"	*.,'?!\-()@/: _;+&%=<> £ \$¥¤[]{}~ ^j¿§#"	*.,'?!\-()@/: _;+&%=<> £ \$¥¤[]{}~ ^j¿§#"	*.,'?!\-()@/: _;+&%=<> £ \$¥¤[]{}~ ^j¿§#"	.*:/@[]
# send	#	#	#	#	#

To enter or edit data:

Do one of the following:

If you want to	Then you can
Enter only digits (1), uppercase (A) characters, lowercase (a) characters, or alphanumeric (2aB) characters.	 Press a keypad key one or more times (depending what input mode you're in) to enter the characters that is displayed on the keypad key. You can press the abc soft key one or more times to switch among uppercase (ABC soft key), numeric (123 soft key), alphanumeric (2aB soft key), uppercase and lowercase (Abc soft key) and lowercase (abc soft key) input modes. For example, if the input mode is ABC: To enter "A", press 2 To enter "B", press 2 twice quickly. To enter "C", press 2 To enter "ZÄÆÅÀÁÂÃÇ", press 2 more than three times quickly. Note: When you are in the uppercase (ABC soft key) or lowercase (abc soft key) input mode, is not available.
Enter special characters.	 Press the keypad key # or *. , or press 0 . For 0 Key: If it is in the uppercase (ABC soft key), uppercase and lowercase (Abc soft key) or lowercase (abc soft key) input mode, it will provide the space character. If it is in the numeric (123 soft key) or

If you want to	Then you can		
	alphanumeric (2aB soft key) input mode, it		
	will only provide the digit 0 .		
	For # www key:		
	- It only provides the pound character # .		
	For \star. key:		
	- If it is in the uppercase (ABC soft key),		
	lowercase (abc soft key), uppercase and		
	lowercase (Abc soft key) or alphanumeric		
	(2aB soft key) input mode, it will provide the		
	following special characters:		
	*.,?!\-()@/:_;+&%=<>£\$¥¤[]{}~^i¿\$#" .		
	- If it is in the numeric (123 soft key) input		
	mode, it will provide the following special		
	characters: .*:/@[].		
	Press () or () to position the cursor to the		
Delete text you entered	right of the text you want to delete, and then press		
Delete text you entered.	the Delete soft key to delete one character at a		
	time.		

Customizing Your Phone

You can customize your SIP-T46G IP phone by personally configuring certain settings, for example, contrast, time & date and ring tones. You can add contacts to the phone's local directory manually or from call history. You can also personalize different ring tones for different callers.

This chapter provides basic operating instructions for customizing your phone. Topics include:

- General Settings
- Audio Settings
- Contact Management
- Call History Management
- Search Source List in Dialing
- System Customizations

If you require additional information or assistance with your new phone, contact your system administrator.

General Settings

Wallpaper

You can customize the wallpaper of the SIP-T46G IP phone, and can change the wallpaper image via phone user interface or web user interface. You can also upload custom pictures as wallpaper images via web user interface.

The SIP-T46G IP phone supports the file format of pictures in the following table:

Format Resolution		File Size		
.jpg/.png/*.bmp	\leqslant 1.8 megapixels	2MB of space should be reserved for the phone		

To upload the custom picture via web user interface:

- 1. Click on Settings->Preference.
- 2. In the **Upload Wallpaper(480*272)** field, click **Browse** to select the desired picture from your local system.

						Log Out English(English)
Yealink T46G	Status Accou	nt Network	DSSKey	Features	Settings	Directory Security
Preference Time & Date Call Display Upgrade Auto Provision Configuration	Live Dialpad Inter Digit Timei Backlight Inactiv Backlight Active Backlight Time(s WatchDog Ring Type	Level econds)	Enabled 4 Low 8 Always On Disabled Ring3.wav	- 0 - 0 - 0 - 0 - 0 - 0 - 0 - 0 - 0		NOTE Selects a language for the web user interface. Live Dialpad It allows P phones to automatically dal out the entered phone number after a specified period of time. Backlight
Dial Plan Voice	Upload Ringtone Wallpaper	ł	Browse No Upload Default.jpg	Cancel	0	Specifies the brightness of the LCD screen display. Contrast Specifies the contrast of the
Ring Tones	Upload Wallpape	r(480*272)	Browse No	o file selected. Cancel	0	LCD screen display. Ring Tones A ring tone that will alert you when a call comes in for the IP
Softkey Layout TR069	Screensaver Wa ScreenSaver Dis Screensaver Typ	play Clock	Never Enabled System	· 0 · 0		phone. Pou can click here to get more guides.
Voice Monitoring		Confirm		Cancel		

3. Click **Upload** to upload the file.

The custom picture appears in the pull-down list of Wallpaper.

You can only delete the custom wallpaper by clicking **Del** when selecting the desired custom wallpaper in the **Wallpaper** field.

Note The SIP-T46G IP phone screen size is 480*272 pixels. If the size of the custom wallpaper doesn't meet the IP phone screen size, the wallpaper will be stretched or zoomed out according to the IP phone screen size.

To change the wallpaper via phone user interface:

- 1. Press Menu->Basic->Display->Wallpaper.
- **2.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select the desired wallpaper image.



3. Press the Save soft key to accept the change or the Back soft key to cancel.

Wallpaper is configurable via web user interface at the path Settings->Preference.

Screen Saver

The screen saver will automatically start when the IP phone is idle for the preset waiting time. The screen saver is used to blank the screen or fill it with moving images or patterns.

The screen saver will be stopped if a phone event occurs:

- Press any key.
- Pick up the handset.
- Hang up the handset.
- There is an incoming call.
- A new prompt (e.g., missed call, new voice mail or forwarded call).
- The status of BLF key changes.

If your phone is idle again for a specified period of time, the screen saver will start again. The time & date and certain feature status icons (e.g., a new text message, auto answer, DND) will also display on the screen saver. For more information on the icons, refer to Icon Instructions on page 3. You can configure the phone whether to display the time & date on the screen saver or not.



You can configure the screen saver of SIP-T46G IP phone, and upload custom pictures to set up a screen saver of your picture via web user interface. You can configure the screen saver wait time, screen saver type and screen saver display clock via phone user interface or web user interface.

The SIP-T46G IP phone supports the file format of custom pictures in the following table:

Format	Resolution	File Size
.jpg/.png/*.bmp	\leqslant 1.8 megapixels	2MB of space should be reserved for the phone

Uploading the Custom Picture via Web User Interface

To upload a custom picture via web user interface:

- **1.** Click on **Settings->Preference**.
- 2. Select Custom from the pull-down list of Screensaver Type.
- **3.** In the **Upload Screensaver** field, click **Browse** to locate the custom picture from your local system, and then click **Upload**.

This field appears only if Screensaver Type is set to Custom.

alink 146G						_		English(Eng	lish)
	Status	Account	Network	DSSKey	Feature	es	Settings	Directory	curity
Preference	Live	Dialpad		Enabled	•	0		NOTE	
Time & Date		r Digit Time(1~14) klight Inactive Leve		4 Low		0		Language Selects a language fo	r the we
Call Display		klight Active Level		8	•	0		user interface.	a che we
Upgrade	Back	klight Time(second	s)	Always On	•	0		It allows IP phones to automatically dial out	
Auto Provision	Wat	chDog		Disabled	•	0		entered phone numb specified period of tir	er after i
Configuration	-	Туре		Ring3.wav	▼	0		Backlight Specifies the brightne	
Dial Plan	Upid	ad Ringtone		Browse N Upload	Cancel		0	LCD screen display.	255 OF CITE
Voice	Wal	paper		Default.jpg	•	0		Specifies the contrast LCD screen display.	t of the
Ring	Upla	ad Wallpaper(480*	*272)	Browse N	lo file selected	ł.	0	Ring Tones	
Tones	6	ensaver Wait Time		Upload	Cancel	~		A ring tone that will a when a call comes in phone.	
Softkey Layout		ensaver wait Time enSaver Display Cl		Enabled	• •	0		You can click her	e to get
TR069	Scre	ensaver Type		Custom	•	0		more guides.	
Voice Monitoring	Scre	ensaver			-	0			
SIP	Upla	ad Screensaver		Browse N	lo file selected	i.	0		
Power Saving				Upload	Cancel				

The custom picture appears in **Screensaver** list. You can only delete the custom picture by clicking **Del** in the **Screensaver** field.

Note You can only upload one custom image at a time via web user interface.

Configuring the Screen Saver

The screen saver wait time can be configured from the following options:

- Never: Screen saver is deactivated.
- **15s**, **30s**, **1min**, **2min**, **5min**, **10min** or **30min**: The screen saver will automatically start each time your phone is idle for a designated amount of time.

To configure the screen saver via phone user interface:

- 1. Press Menu->Basic->Display->Screensaver.
- 2. Press (•) or (•), or the Switch soft key to select the desired wait time from the Wait Time field.
- **3.** Press (•) or (•), or the **Switch** soft key to select the desired value from the **Display Clock** field.

4. Press (•) or (•), or the **Switch** soft key to select the desired screen saver type from the **Screensaver Type** field.

This field is configurable only if you have uploaded a custom picture via web user interface.

Sc	reensaver	
1. Wait Time:	15s	< ⊳
2. Display Clock:	Enabled	< ⊳
3. Screensaver Type:	System	< >
Back	Switch	Save

5. Press the **Save** soft key to accept the change or the **Back** soft key to cancel.

To configure the screen saver via web user interface:

- 1. Click on Settings->Preference.
- 2. Select the desired wait time from the pull-down list of Screensaver Wait Time.
- 3. Select the desired value from the pull-down list of ScreenSaver Display Clock.
- 4. Select the desired screen saver type from the pull-down list of Screensaver Type.
 - If you select System.

The IP phone will automatically set the built-in picture as the screen saver.

Yealink 1466				Log Out English(English) v
	Status Account Netwo	ork DSSKey Features	Settings	Directory Security
Preference Time & Date Call Display Upgrade Auto Provision Configuration Dial Plan Voice	Live Dalpad Live Dalpad Inter Digit Time(1~145) Backlight Inactive Level Backlight Active Level Backlight Active Level Backlight Time(seconds) WatchDog Ring Type Upload Ringtone Wallpaper	Enabled • ? 4 ? Low • ? 8 • ? Always On • ? Disabled • ? Rng3.wav • ? Browse No file selected. Upload Cancel Default.jpg • ?	Q	NOTE Selects a language for the web user interface. Lie Books IP phones to automatically dai out the entered phone number after a specified period of time. Backlight Specifies the bightness of the LCD screen display. Contrast Specifies the contrast of the
Ring	Upload Wallpaper(480*272)	Browse No file selected.	0	LCD screen display. Ring Tones A ring tone that will alert you when a call comes in for the IP
Softkey Layout	Screensaver Walt Time ScreenSaver Display Clock Screensaver Type	15s • 🖓 Enabled • System • 🖓		 Phone. You can click here to get more guides.
Voice Monitoring	Confirm	Cancel		

- If you select **Custom**.

The IP phone will automatically set the custom pictures you upload as the screen saver, and display these pictures alternately.

5. Click **Confirm** to accept the change.

Power Saving

The power saving feature is used to turn off the backlight and screen to conserve energy. The IP phone enters power-saving mode after it has been idle for a certain period of time.

The IP phone will exit power-saving mode if a phone event occurs:

- Press any key.
- Pick up the handset.
- Hang up the handset.
- There is an incoming call.
- A new prompt (e.g., missed call, new voice mail or forwarded call).
- The status of BLF key changes.

If the screen saver is enabled on your phone, power-saving mode will still occur.

You can configure the following power-saving settings:

- Office Hour
- Idle Timeout (minutes)

Note Power saving is configurable via web user interface only.

Enabling the Power Saving via Web User Interface

The office hour and idle timeout (minutes) settings work only if the power saving feature is enabled.

To enable the power saving feature via web user interface:

- 1. Click on Settings->Power Saving.
- 2. Select Enabled from the pull-down list of Power Saving.

Yealink 1466	_						Eng	Log Out Jish(English) 🗸
	Status	Account	Network	DSSKey	Features	Settings	Directory	Security
Preference	Pov	wer Saving		Enabled	• 0		NOTE	
Time & Date		fice Hour 🕜		07 - 19			Settings Pov	versaving
Call Display		Tuesday		07 - 19			You can cl more guides.	ick here to get
Upgrade		Wednesday		07 - 19			J. J	
Auto Provision		Thursday		07 - 19				
Configuration		Friday		07 19				
Dial Plan		Saturday Sunday		07 07				
Voice		e TimeOut (minu	tes)					
Ring		Office Hour Idle Ti	imeOut	360	0			
Tones		Off Hour Idle Time		10	0			
Softkey Layout		User Input Extens	ion Idle TimeOut	10	0			
TR069		Confi	irm		Cancel			
Voice Monitoring								
SIP								
Power Saving								

3. Click **Confirm** to accept the change.

Configuring the Office Hour via Web User Interface

Office Hour specifies the starting time and ending time in the office each day.

To configure the office hour via web user interface:

- **1.** Click on **Settings**->**Power Saving**.
- **2.** Select a desired day of the week.

3. Enter the starting time and ending time respectively in the desired day field.

	Status	Account	Network	DSSKey	Features	Settings	Directory Security
Preference	Po	wer Saving		Enabled	• 0		NOTE
Fime & Date	Of	fice Hour 🛛 🕜					Settings Powersaving
		Monday		07 - 19			
Call Display		Tuesday		07 - 19			You can click here to get more guides.
Jpgrade		Wednesday		07 - 19			, i i i i i i i i i i i i i i i i i i i
Auto Provision		Thursday		07 - 19			
Configuration		Friday		07 19			
Configuration		Saturday		07 - 07			
Dial Plan		Sunday		07 07			
/oice	Id	le TimeOut (minu	tes)				
Ring		Office Hour Idle Ti	meOut	360	0		
Tones		Off Hour Idle Time	Out	10	0		
		User Input Extensi	on Idle TimeOut	10	0		
Softkey Layout		Confi			Cancel		
TR069		Com			Calicer		
Voice Monitoring							
SIP							

4. Click **Confirm** to accept the change.

Configuring the Idle Timeout via Web User Interface

Idle Timeout specifies the period of time before the IP phone enters power-saving mode. The following three types of idle timeout you can configure:

- Office Hours Idle TimeOut: Configures the idle timeout for office hours.
- Off Hours Idle TimeOut: Configures the idle timeout for non-office hours.
- **User Input Extension Idle TimeOut**: Configures idle timeout that applies after you use the IP phone (for example, press a key on the phone or pick up/hang up the handset).

By default, the Office Hours Idle Timeout is much longer than the Off Hours Idle TimeOut. If you use the IP phone, the idle timeout that applies (User Input Extension Idle TimeOut or Office Hours/Off Hours Idle TimeOut) is the timeout with the highest value. If the phone has an incoming call or new message, the User Input Extension Idle TimeOut is ignored.

To configure the idle timeout via web user interface:

- 1. Click on Settings->Power Saving.
- Enter the desired value in the Office Hours Idle TimeOut field. The default value is 360, you can set to 1-600.
- Enter the desired value in the Off Hours Idle TimeOut field.
 The default value is 10, you can set to 1-10.
- 4. Enter the desired value in the User Input Extension Idle TimeOut field.

alink 1466	Status	Account	Network	DSSKey	Features	Settings	Directory Security
Preference	Pc	wer Saving		Enabled	• 0		NOTE
Fime & Date	01	fice Hour 🕜					Settings Powersaving
Call Display		Monday		07 19			You can click here to get
Cali Display		Tuesday		07 - 19			more guides.
Upgrade		Wednesday		07 19			
Auto Provision		Thursday		07 - 19			
Configuration		Friday		07 - 19			
-		Saturday		07 - 07			
Dial Plan		Sunday		07 07			
Voice	Id	le TimeOut (minu	tes)				
Ring		Office Hour Idle Ti	meOut	360	0		
Tones		Off Hour Idle Time	Out	10	0		
		User Input Extensi	on Idle TimeOut	10	0		
Softkey Layout		Confi	m	ſ	Cancel		
TR069		Com			Cancer		
Voice Monitoring							
STP							

The default value is 10, you can set to 1-30.

5. Click **Confirm** to accept the change.

Backlight

Backlight has three options:

Active Level: The intensity of the LCD screen when the phone is active. You can change the intensity of the LCD screen. Digits (1-10) represent different intensities. 10 is the highest intensity.

Inactive Level: The intensity of the LCD screen when the phone is inactive. You can select a low intensity or turn off the backlight.

Backlight Time: The delay time to change the intensity of the LCD screen when the phone is inactive. You can select a desired time to change the intensity or turn on the backlight permanently.

- Always On: Backlight is on permanently.
- **15s**, **30s**, **1min**, **2min**, **5min**, **10min** or **30min**: Backlight is changed when the phone is inactive after the designated time (in seconds).

You can also change the intensity of the LCD screen of EXP40 connected to the SIP-T46G IP phone.

To configure the backlight via phone user interface:

- 1. Press Menu->Basic->Display->Backlight.
- **2.** Press (•) or (•), or the **Switch** soft key to select the desired value from the **Active Level** field.
- 3. Press () or (), or the Switch soft key to select the desired value from the Inactive

Level field.

4. Press () or () , or the **Switch** soft key to select the desired time from the **Backlight Time** field.

1. Active Level:	8	$\triangleleft \triangleright$
2. Inactive Level:	Low	< >
3. Backlight Time:	Always On	$\triangleleft \triangleright$

5. Press the Save soft key to accept the change or the Back soft key to cancel.

Backlight is configurable via web user interface at the path Settings->Preference.

Contrast

You can only configure the contrast of the expansion module EXP40 connected to the SIP-T46G IP phone to a comfortable level on the phone. Ensure that the expansion module has been connected to the phone before configuration. The intensity of contrast ranges from 1 to 10 and the highest intensity is 10.

To configure the contrast via phone user interface:

1. Press Menu->Basic->Display->Contrast.

If EXP40 is not connected to the phone, the Contrast Setting screen displays "No EXP".

Press (•) or (•), or the Switch soft key to increase or decrease the intensity of contrast.
 The default contrast level is "6".



3. Press the Save soft key to accept the change or the Back soft key to cancel.

Language

The default language of the phone user interface is English. If the language of your web browser

is not supported by the phone, the web user interface will use English by default. You can change the language for the phone user interface and the web user interface respectively.

To change the language for the phone user interface directly:

- 1. Press Menu->Basic->Language.
- **2.** Press (\bullet) or (\bullet) to select the desired language.



3. Press the Save soft key to accept the change.

Text displayed on the phone user interface will change to the selected language.

To change the language for the web user interface:

1. Select the desired language from the pull-down list at the top-right corner of web user interface.

Yealink 1466	Status Account Networ	k Dsskey Features	Log Out English(English) - Settings Directory Security
Status	Version 🕜		NOTE
	Firmware Version	28.81.0.20	
	Hardware Version	28.0.0.128.0.0.0	Version It shows the version of firmware
	Device Certificate 🕜		and hardware.
	Device Certificate	Factory Installed	Network It shows the network settings of Internet (WAN) port.
	Network 🕜		
	Internet Port	IPv4	Account It shows the registration status of SIP accounts.

Text displayed on the web user interface will change to the selected language.

Time & Date

The time and date are displayed on the LCD screen when the phone is idle. You can configure the phone to obtain the time and date from the SNTP server automatically, or configure the time and date manually. If the phone cannot obtain the time and date from the Simple Network Time Protocol (SNTP) server, contact your system administrator for more information.

To configure the SNTP settings via phone user interface:

- 1. Press Menu->Basic->Time & Date->General->SNTP Settings.
- 2. Press (\cdot) or (\cdot) , or the Switch soft key to select the time zone that applies to your area

from the Time Zone field.

The default time zone is "+8".

- Enter the domain name or IP address of SNTP server in the NTP Server1 and NTP Server2 field respectively.
- **4.** Press (•) or (•), or the **Switch** soft key to select the desired value from the **Daylight Saving** field.
- **5.** Press (•) or (•), or the **Switch** soft key to select the desired time zone name from the **Location** field.

The default time zone name is "China(Beijing)".

	General	
1. Time Zone:	+8	< ►
2. NTP Server1:	cn.pool.ntp.org	
3. NTP Server2:	pool.ntp.org	
4. Daylight Saving:	Automatic	<►
5. Location:	China(Beijing)	
Back	Switch	Save

6. Press the Save soft key to accept the change or the Back soft key to cancel.

Note Please refer to Appendix A - Time Zones for the list of available time zones on the IP phone.

To configure the time and date manually via phone user interface:

- 1. Press Menu->Basic->Time & Date->General->Manual Settings.
- 2. Enter the specific date and time.

2016-10-10 16:07:	09 🌲	
2016-10-10 16:07:09		
	Save	

3. Press the Save soft key to accept the change.

The time and date displayed on the LCD screen will change accordingly.

To configure the time and date format via phone user interface:

- 1. Press Menu->Basic->Time & Date->Time & Date Format.
- 2. Press (\cdot) or (\cdot), or the **Switch** soft key to select the desired date format from the **Date**

Format field.

Press (•) or (•), or the Switch soft key to select the desired time format (12 Hour or 24 Hour) from the Time Format field.

Time	e & Date Format	
1. Date Format:	WWW MMM DD	$\triangleleft \triangleright$
2. Time Format:	24 Hour	$\triangleleft \triangleright$
Back	Switch	Save

4. Press the **Save** soft key to accept the change or the **Back** soft key to cancel.

There are 7 available date formats. For example, for the date format "WWW DD MMM", "WWW" represents the abbreviation of the weekday, "DD" represents the two-digit day, and "MMM" represents the first three letters of the month.

The date formats available:

Date Format	Example (2016-09-02)	
WWW MMM DD	Fri, Sep 02	
DD-MMM-YY	02-Sep-16	
YYYY-MM-DD	2016-09-02	
DD/MM/YYYY	02/09/2016	
MM/DD/YY	09/02/16	
DD MMM YYYY	02 Sep, 2016	
WWW DD MMM	Fri, 02 Sep	

Time and date are configurable via web user interface at the path **Settings**->**Time & Date**.

Note

You can also customize the date format. Contact your system administrator for more information.

Administrator Password

The Advanced Settings option is only accessible to the administrator. The default administrator password is "admin". For security reasons, you should change the default administrator password as soon as possible.

To change the administrator password via phone user interface:

- 1. Press Menu->Advanced (default password: admin) ->Change Password.
- 2. Enter the old password in the Old PWD field.
- 3. Enter the new password in the New PWD field.
- 4. Re-enter the new password in the Confirm PWD field.

Change Password				
1. Old PWD: ••••				
2. New PWD:				
3. Confirm PWD:				
Back	abc	Delete	Save	

5. Press the Save soft key to accept the change or the Back soft key to cancel.

Administrator password is configurable via web user interface at the path Security->Password.

Key As Send

You can set the "#" key or "*" key to perform as a send key while dialing.

To configure key as send via phone user interface:

- 1. Press Menu->Features->Others->General.
- 2. Press (•) or (•), or the Switch soft key to select # or * from the Key As Send field, or select Disabled to disable this feature.

General			
1. Key As Send:	#	$\triangleleft \triangleright$	
2. History Record:	Enabled	< ⊳	
Back	Switch	Save	

3. Press the Save soft key to accept the change or the Back soft key to cancel.

Key as send is configurable via web user interface at the path Features->General Information.

Phone Lock

You can lock your phone temporarily when you are not using it. This feature helps to protect your phone from unauthorized use.

Phone lock consists of the following:

Menu Key:	The Menu soft key is locked. You cannot access the menu of the
	phone until unlocked.
Function Keys:	The function keys are locked. You cannot use the MESSAGE, REDIAL,
	HOLD, MUTE, TRANSFER, OK, X, navigation keys and soft keys until
	unlocked.
All Keys:	All keys are locked except the Volume key, digit keys, # key, * key and
	Speakerphone key. You are only allowed to dial emergency numbers,
	reject incoming calls by pressing the X key or Reject soft key, answer
	incoming calls by lifting the handset, pressing the Speakerphone key,
	the HEADSET key, the OK key or Answer soft key, and end the call by
	hanging up the handset, pressing the Speakerphone key, End Call
	soft key or X key.

Note The emergency number setting, if desired, must be made before lock activation. For more information, refer to Emergency Number on page 113.

To activate the phone lock via phone user interface:

- 1. Press Menu->Basic->Phone Lock.
- Enter the desired PIN (default PIN: 123) in the Unlock PIN field, and then press the OK soft key.
- **3.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select **Enabled** from the **Lock Enable** field.
- **4.** Press (•) or (•), or the **Switch** soft key to select the desired type from the **Lock Type** field.
- 5. Enter the desired interval of automatic phone lock in the Lock Time Out field.

The default timeout is 0. It means the phone will not be automatically locked. You need to long press $\boxed{\#_{mn}}$ to lock it immediately when the phone is idle.

If it is set to other values except 0 (e.g., 5), the phone will be locked when the phone is inactive in idle screen for the designated time (in seconds).

	Phor	ne Lock	
1. Lock Enable:		Enabled	∢ ►
2. Lock Type:		All Keys	<►
3. Lock Time Out:		0	
Back	123	Delete	Save

6. Press the **Save** soft key to accept the change.

When the phone is locked, the LCD screen prompts "Phone locked." and displays the icon $\fbox{}$.

1006	l	14	4:49 Fri, Sep 02
1006			
🕜 Not	te		
	Phone I	ocked.	
History	Directory	DND	Menu

To change the phone unlock PIN via phone user interface:

- 1. Press Menu->Basic->Change PIN.
- 2. Enter the desired value in the Old PIN, New PIN and Confirm PIN field respectively.

	Char	nge PIN	
1. Old PIN:		•••	
2. New PIN:		••••	
3. Confirm PIN:			
Back	123	Delete	Save

3. Press the Save soft key to accept the setting or the Back soft key to cancel.

Note The unlock PIN length must be within 15 digits.

To unlock the phone via phone user interface:

- 1. Press any locked key, the LCD screen prompts "Unlock PIN".
- 2. Enter the desired PIN in the Unlock PIN field.

<u>ຈ</u> 1006		11	. 22 Fri, Sep 02
1006			
1 Phor	ne Lock		
Unlo	ck PIN 🐽		
Cancel	123	Delete	ОК

3. Press the **OK** soft key to unlock the phone.

The icon 🔒 disappears from the LCD screen.

You can long press $[#_{mod}]$ or wait for a period of time (if configured) to lock the phone again.

Note You can also unlock the phone by administrator password. When you enter the administrator password to unlock the phone, the phone will turn to the Change PIN screen.

To deactivate the phone lock via phone user interface:

- 1. Press Menu->Basic->Phone Lock.
- Enter the desired PIN (default PIN: 123) in the Unlock PIN field, and then press the OK soft key.
- **3.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select **Disabled** from the **Lock Enable** field.

Phone Lock				
1. Lock Enable: Disabled < >				
2. Lock Type:	All Keys	< >		
3. Lock Time Out: 0				
Back	Switch	Save		

4. Press the Save soft key to accept the change.

Phone lock is configurable via web user interface at the path Features->Phone Lock.

Audio Settings

Volume

You can press the Volume key to adjust the ringer volume when the phone is idle or ringing. You can also press the Volume key to adjust the receiver volume of currently engaged audio devices (handset, speakerphone or headset) when the phone is in use.

To adjust the ringer volume:

When the phone is idle:

1. Press **—** to adjust the ringer volume.



When the phone is ringing:

1. Press - to adjust the ringer volume.



You can also press to adjust the ringer volume when selecting a ring tone. For more information, refer to Ring Tones on page 57.

Note

If ringer volume is adjusted to minimum, the icon 🗱 will appear on the LCD screen.

To adjust the volume when the phone is during a call:

1. Press **—** to adjust the volume of currently engaged audio device (handset, speakerphone or headset).



You can also press to adjust the volume when playing back the recording calls. For more information, refer to Playing Back Recorded Calls on page 175.

Ring Tones

Ring tones are used to indicate incoming calls. You can select different ring tones to distinguish different accounts registered on your phone, or to distinguish your phone from your neighbor's.

To select a ring tone for the phone via phone user interface:

- 1. Press Menu->Basic->Sound->Ring Tones->Common.
- **2.** Press (\bullet) or (\bullet) to select the desired ring tone.
- 3. (Optional.) Press to adjust the ringer volume.

Ring Tones				
• Ring1.wav				
O Ring2.wav				
Ring3.wav				
Ring4.wav				
O Ring5.wav				
Back	Save			

4. Press the Save soft key to accept the change or the Back soft key to cancel.

A ring tone for the phone is configurable via web user interface at the path **Settings->Preference->Ring Type**.

To select a ring tone for the account via phone user interface:

- 1. Press Menu->Basic->Sound->Ring Tones.
- 2. Press () or () to select the desired account and then press the **Enter** soft key.

Ring Tones		
1. Common		
2. 1006		

3. Press (\bullet) or (\bullet) to select the desired ring tone.

If **Common** is selected, this account will use the ring tone selected for the phone.

Ring Tones				
• Common				
O Ring1.wav				
O Ring2.wav				
C Ring3.wav				
O Ring4.wav				
Back	Save			

- **4.** (Optional.) Press to adjust the ringer volume.
- 5. Press the Save soft key to accept the change or the Back soft key to cancel.

A ring tone for the account is configurable via web user interface at the path **Account->Basic->Ring Type**.

To upload a custom ring tone for your phone via web user interface:

- 1. Click on Settings->Preference.
- 2. In the **Upload Ringtone** field, click **Browse** to locate a ring tone (the file format must be *.wav) file from your local system.

Yealink 1466								Eng	Log Out lish(English) 🗸
TEAIINK T46G	Status	Account	Network	DSSKey	Featu	res	Settings	Directory	Security
Preference Time & Date Call Display Upgrade	Inter Back Back Back	Dialpad r Digit Time(1~14s) dight Inactive Level dight Active Level dight Time(seconds		Enabled 4 Low 8 Always On	•	0 0 0 0		user interface. Live Dialpad It allows IP ph automatically d	ones to ial out the
Auto Provision Configuration Dial Plan	Ring	chDog Type ad Ringtone		Disabled Ring3.wav Browse Upload	▼ ▼ No file selecte Cancel	2 2 ed.	0	specified perio	rightness of the
Voice Ring Tones	Uplo	paper ad Wallpaper(480*.	272)	Upload	▼ No file selecte Cancel]	0	Specifies the c LCD screen dis Ring Tones A ring tone the when a call co	
Softkey Layout TR069 Voice Monitoring	Scre	ensaver Wait Time enSaver Display Clo ensaver Type Confir		15s Enabled System	▼ ▼ Cancel	0		phone. Prou can cl more guides.	ck here to get

3. Click Upload to upload the file.

The custom ring tone appears in the pull-down list of **Ring Type**.

You can only delete the custom ring tone by clicking **Del** when selecting the desired custom ring tone in the **Ring Type** field.

The priority of ring tone for an incoming call on the phone is as follows: Contact ring tone (refer to Adding Contacts) >Group ring tone (refer to Adding Groups) >Account ring tone >Phone ring tone.

Single custom ring tone file must be within 8MB and total custom ring tone files must be within 20MB.

Uploading custom ring tones for your phone is configurable via web user interface only.

Key Tone

Note

If you enable key tone, the phone will produce a sound when you press the keypad.

To configure key tone via phone user interface:

- 1. Press Menu->Basic->Sound->Key Tone.
- 2. Press (•) or (•), or the Switch soft key to select Enabled or Disabled from the Key Tone field.

Key Tone				
1. Key Tone: Enabled < >				
ac.				
Back	Switch	Save		

3. Press the Save soft key to accept the change or the Back soft key to cancel.

Key tone is configurable via web user interface at the path Features->Audio.

BLF Ring Type

You can specify a desired ring tone to play for BLF pickup on the phone. This allows the monitoring phone to play a specific ring tone when the monitored line receives an incoming call. To use this feature, you need to enable visual and audio alert features in advance. For more information, refer to Configuring Visual and Audio Alert for BLF pickup on page 164.

To configure BLF ring type via phone user interface:

- 1. Press Menu->Basic->Sound->BLF Ring Type.
- **2.** Press (\bullet) or (\bullet) to select the desired ring tone.

Ring Tones				
• Ring1.wav				
O Ring2.wav				
O Ring3.wav				
O Ring4.wav				
O Ring5.wav				
Back	Save			

- 3. (Optional.) Press to adjust the ringer volume.
- 4. Press the **Save** soft key to accept the change or the **Back** soft key to cancel.

BLF ring type is configurable via web user interface at the path Features->Call Pickup.

Contact Management

This section provides the operating instructions for managing contacts. Topics include:

- Directory
- Local Directory
- Blacklist
- Remote Phone Book

Directory

Directory provides easy access to frequently used lists. The lists may contain Local Directory, History, Remote Phone Book and LDAP. You can configure the list(s) to access for the **Directory** soft key.

To configure the list(s) to access for the Directory soft key via web user interface:

- 1. Click on Directory->Setting.
- In the **Directory** block, select the desired list from the **Disabled** column and then click →.

The selected list appears in the **Enabled** column.

- 3. Repeat the step 2 to add more lists to the **Enabled** column.
- 4. To remove a list from the **Enabled** column, select the desired list and then click _____.
- To adjust the display order of enabled lists, select the desired list and then click
 or

 .

The LCD screen will display the list(s) in the adjusted order.

Yealink 1466		Log Out English(English) -
	Status Account Network DSSKey Features Settings	Directory Security
Local Directory Remote Phone Book Phone Call Info LDAP Multicast IP Setting	Directory Disabled Enabled Local Directory History	NOTE Directory It provides easy access to frequently used lets. Search Source in Dialing It alows the IP phone to automatically search entries from the search source list based on the entered string, and display results on the pre-dailing screen. Recent Call In Dialing It alows users to view the placed calls twhen the phone
	Search Source List In Dialing 💡	is on the pre-dialing screen.
	Disabled Enabled Local Directory	You can dick here to get more guides.
	Recent Call In Dialing Enabled 🔹 💡	
	Confirm	

6. Click **Confirm** to accept the change.

Note The list(s) to access for **Directory** soft key is configurable via web user interface only.

To view the directory via phone user interface:

1. Press the **Directory** soft key when the phone is idle.

The LCD screen displays the enabled list(s) in the directory.

Directory				
1. Local Directory				
2. History				
Cancel	Enter			

If there is only one list in the directory, press the **Directory** soft key to enter this list directly.

Note If the remote phone book and LDAP are not configured in advance, you cannot see remote phone book and LDAP lists on the phone user interface. For more information on remote phone book, refer to Remote Phone Book on page 77. For more information on LDAP, contact your system administrator.

Local Directory

The built-in phone directory can store the names and phone numbers of your contacts. You can store up to 1000 contacts and 48 groups in your phone's local directory. You can add new groups and contacts, edit, delete or search for a contact, or simply dial a contact number from the local directory.

Local directory can be backed up on the provisioning server. For more information, contact your system administrator.

Adding Groups

Note

To add a group to the local directory:

Local Directory

 1. All Contacts

 Back
 AddGroup
 Search
 Enter

The IP phone enters the local directory directly as there is only Local Directory enabled in the directory by default.

If Local Directory is removed from the directory (refer to Directory on page 61), press **Menu->Directory->Local Directory** to enter the local directory.

- 2. Press the AddGroup soft key.
- 3. Enter the desired group name in the Name field.
- **4.** Press () or () , or the **Switch** soft key to select the desired group ring tone from the **Ring** field.

If **Auto** is selected, this group will use the ring tone according to the priority: Contact ring tone (refer to Adding Contacts) >Account ring tone (refer to Ring Tones) >Phone ring tone (refer to Ring Tones). If a specific ring tone is selected, this group will use the ring tone according to the priority: Contact ring tone (refer to Adding Contacts) >Group ring tone.

Add Group				
1. Name: Test				
2. Ring Tones: Ring1.wav <				
Back	Switch	Save		

5. Press the Save soft key to accept the change or the Back soft key to cancel.

You can also edit or delete any newly added contact groups.

Editing Groups

To edit a group in the local directory:

The IP phone enters the local directory directly as there is only Local Directory enabled in the directory by default.

Local Directory					
1. All Contacts					
2. Test	2. Test				
Back	AddGroup	Option	Enter		

If Local Directory is removed from the directory (refer to Directory on page 61), press Menu->Directory->Local Directory to enter the local directory.

- 2. Select the desired contact group.
- 3. Press the **Option** soft key, and then select **Detail** from the prompt list.

	Local Directory	
1. All Cont	acts	
2. Test	Detail	
Delete		
	Delete All	
Cancel		ОК

- 4. Press (\bullet) or (\bullet) to highlight the group information and then edit.
- 5. Press the Save soft key to accept the change or the Back soft key to cancel.

Deleting Groups

To delete a group from the local directory:

The IP phone enters the local directory directly as there is only Local Directory enabled in the directory by default.

Local Directory				
1. All Contac	ts			
2. Test				
Back	AddGroup	Option	Enter	

If Local Directory is removed from the directory (refer to Directory on page 61), press Menu->Directory->Local Directory to enter the local directory.

- 2. Select the desired contact group.
- 3. Press the Option soft key, and then select Delete from the prompt list.

The LCD screen prompts the following warning:

	ocal Directory
1. All Contacts	
2. Tes 🕕 Warning	
Dele	e selected group?
Cancel	ОК

4. Press the **OK** soft key to confirm the deletion or the **Cancel** soft key to cancel.

You can also delete all groups by pressing the **Option** soft key, and then select **Delete All**.

Adding Contacts

You can add contacts to the local directory in one of the following ways:

- Manually
- From call history
- From a remote phone book

Adding Contacts Manually

To add a contact to the local directory manually:

1. Press the **Directory** soft key.

The IP phone enters the local directory directly as there is only Local Directory enabled in

the directory by default.

	Local D	virectory	
1. All Contac	ts		
	1		C
Back	AddGroup	Search	Enter

If Local Directory is removed from the directory (refer to Directory on page 61), press Menu->Directory->Local Directory to enter the local directory.

2. Select the desired contact group and then press the Enter soft key.

If the contact you want to add doesn't belong to any group, select **All Contacts** and then press the **Enter** soft key.

- 3. Press the Add soft key.
- 4. Enter the name and the office, mobile or other numbers in the corresponding fields.

Ad	dd Contact	
1. Name:	А	
2. Office Number:	123	
3. Mobile Number:	234	
4. Other Number:		
5. Account:	Auto	$\triangleleft \triangleright$
Back	Switch	Save

5. Press (•) or (•), or the **Switch** soft key to select the desired account from the **Account** field.

If **Auto** is selected, the phone will use the default account when placing calls to the contact from the local directory.

6. Press (•) or (•), or the **Switch** soft key to select the desired ring tone from the **Ring** field.

If **Auto** is selected, this contact will use the ring tone according to the priority: Group ring tone (refer to Adding Groups) >Account ring tone (refer to Ring Tones) >Phone ring tone (refer to Ring Tones).

- 7. Press (\cdot) or (\cdot) , or the **Switch** soft key to select the desired group from the **Group** field.
- 8. Press (\bullet) or (\bullet) to select **Photo** field and then press the **Enter** soft key.
- **9.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select the desired photo from the **Photo** field.

10. Press the Save soft key to accept the change or the Back soft key to cancel.

Note If the contact already exists in the directory, the LCD screen will prompt "Contact name existed!".

Adding Contacts from Call History

To add a contact to the local directory from call history:

- 1. Press the **History** soft key.
- **2.** Press (\bullet) or (\bullet) to highlight the desired entry.
- 3. Press the Option soft key, and then select Add to Contact from the prompt list.

	All		Misseo	d Plac	ced	1/5 🕨
-	4607 Yeal	ink		Today	09:58	
1		Detail			09:48	
Į	4607 Yeal	Add to	o Contact		09:48	
1	4607	Add to	o Blacklist	l.	09:47	
4	4603	Delete	e All	1.1.1.0.1	09:47	
		1		roud)	1	
	Cancel				0	K

- 4. Press the **OK** soft key. And then enter the contact name.
- 5. Press the Save soft key to accept the change.

The entry is successfully saved to the local directory.

Adding Contacts from remote phone book

To add a contact to the local directory from remote phone book:

1. Press Menu->Directory->Remote Phone Book.

If Remote Phone Book is added to the directory (refer to Directory on page 61), press **Directory->Remote Phone Book** to enter the remote phone book.

- 2. Select the desired remote group and then press the Enter soft key.
- **3.** Press (\bullet) or (\bullet) to highlight the desired entry.
- 4. Press the Option soft key, and then select Add to Contact from the prompt list.
- 5. Press the Save soft key to save the contact to the local directory.

If the contact already exists in the local directory, the LCD screen will prompt "Overwrite the original contact?". Press the **OK** soft key to overwrite the original contact in the local directory or the **Cancel** soft key to cancel.

For more information on remote phone book operating, refer to Remote Phone Book on page 77.

Editing Contacts

To edit a contact in the local directory:

1. Press the **Directory** soft key.

The IP phone enters the local directory directly as there is only Local Directory enabled in the directory by default.

	Local D	irectory	
1. All Conta	cts		
Back	AddGroup	Search	Enter

If Local Directory is removed from the directory (refer to Directory on page 61), press **Menu->Directory->Local Directory** to enter the local directory.

2. Select the desired contact group and then press the **Enter** soft key.

If the contact is not in any group, select **All Contacts** and then press the **Enter** soft key

- **3.** Press (\bullet) or (\bullet) to highlight the desired contact.
- 4. Press the Option soft key, and then select Detail from the prompt list.
- **5.** Press \bigcirc or \bigcirc to highlight the contact information and then edit.

		A	
1. Name:		A	
2. Office Num	per:	123	
3. Mobile Num	ber:	234	
4. Other Numb	per:		
5. Account:		Auto	<►
Back	Abc	Delete	Save

6. Press the Save soft key to accept the change or the Back soft key to cancel.

Deleting Contacts

To delete a contact from the local directory:

The IP phone enters the local directory directly as there is only Local Directory enabled in the directory by default.

If Local Directory is removed from the directory (refer to Directory on page 61), press **Menu->Directory->Local Directory** to enter the local directory.

Search

Enter

2. Select the desired contact group and then press the Enter soft key.

AddGroup

If the contact is not in any group, select All Contacts and then press the Enter soft key

3. Press (\bullet) or (\bullet) to highlight the desired contact.

Back

4. Press the Option soft key, and then select Delete from the prompt list.

The LCD screen prompts the following warning:

Y	All Contacts	1/1 🕨
<u> </u>	e	123
	() Warning	
	Delete	e selected item?
Ca	ancel	ОК

5. Press the OK soft key to confirm the deletion or the Cancel soft key to cancel.

You can also delete all contacts by pressing the Option soft key, and then select Delete All.

Uploading Contact Photos

To upload a new custom photo for the contact via web user interface:

- 1. Click on Directory->Local Directory.
- 2. Click Browse, and then select a photo from your local computer.
- 3. Click Upload Photo to upload the photo.

adials -								En	Log (glish(English)
alink 1466	Status	Account	Network	DSSKey	Featu	res Set	tings	Directory	Security
Directory	Index	Name	Office Number M	1obile Number Other	Number	All Contacts 👻		NOTE	
,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,	1	Joy	<u>1000</u>	<u>1000 1</u>	000	All Contacts			
ne	2							Local Direct	o ry one directory ca
	3							store the nam	es and phone
ll Info	4							numbers of yo	ur contacts.
	5								ew groups and
	7								, delete or seard or simply dial a
Р	8							contact numbe directory.	er from the local
	9								
	10							You can impor contact list.	t or export the
	Page 1 👻 F	Prev Next	Hang Up	Delete All De	elete	Move To All (Conta: 👻		
								more guides	lick here to ge
	Directory (_					
	Name				-				
	Office Number								
	Mobile Number				A CONTRACTOR	and high			
	Other Number								
	Ring Tone	Auto	•		Delete P	hoto			
	Group		ontacts 👻	Browse	. No file se	elected.			
	Account	Auto		Upload P	hoto				
	Photo		angeas.jpg 👻	Import Local D					
	Add	E	dit		o file selected				
	Group Setting	J 🕜		Import XML	Export XI				
	Group				o file selected				
	Ring	Auto	-	Import CSV	Export CS	SV Show T	itle		
	Add	Edit Delete	Delete All						

Note

The phone only supports *.png, *.jpg and *.bmp format files.

You can only delete custom photos.

To change the custom photo for the contact via web user interface:

- **1.** Click on **Directory**->**Local Directory**.
- 2. Click to select the desired entry which you want to edit.
- 3. Select the desired photo from the pull-down list of **Photo**.

4. Click **Edit** to accept the change.

ealink 1746G					_		En	Log O glish(English)
	Status	Account	Network	DSSKey	Feature	es Settings	Directory	Security
Local Directory	Index	Name	Office Number M	obile Number Other	Number /	All Contacts 👻 📃	NOTE	l
	1	Joy	<u>1000</u>	<u>1000 10</u>	000	All Contacts 🛛 🗸		
Remote Phone Book	2							one directory can
	4						store the name numbers of vo	
Phone Call Info	5						You can add n	
LDAP	6							ew groups and delete or search
	7						for a contact,	or simply dial a r from the local
Multicast IP	8						directory.	r from the local
Setting	9						You can import	or export the
occung	10						contact list.	or export and
	Directory (_			more guides.	lick here to get
	Name	Joy			Care.			
	Office Number	1000						
	Mobile Number	1000						
	Other Number	1000]		N. Martin		
	Ring Tone	Auto	-		Delete Ph	oto		
	Group	All Co	ontacts 👻	Browse	No file sele	ected.		
	Account	Auto	-	Upload P	hoto			
	Photo	Hydra	angeas.jpg 👻	Import Local D	irectory File	• 🕜		
	Add	E	dit		file selected.	_		
	Group Setting	J 🕜		Import XML	Export XML			
	Group				file selected.			
	Ring	Auto	•	Import CSV	Export CS\	Show Title		
	Add	Edit Delete	Delete All					

When you place a call to the contact or receive a call from the contact, the LCD screen will show the contact photo.



Placing Calls to Contacts

To place a call to a contact from the local directory:

The IP phone enters the local directory directly as there is only Local Directory enabled in the directory by default.

	Local Di	rectory	
1. All Conta	cts		

If Local Directory is removed from the directory (refer to Directory on page 61), press Menu->Directory->Local Directory to enter the local directory.

2. Select the desired contact group and then press the Enter soft key.

If the contact is not in any group, select **All Contacts** and then press the **Enter** soft key.

- **3.** Press (\bullet) or (\bullet) to highlight the desired contact.
- **4.** Do one of the following:
 - If only one number for the contact is stored in the local directory, press the **Send** soft key to dial out the number.
 - If multiple numbers for the contact are stored in the local directory, press the **Send** soft key to display a list of numbers.

Press (\bullet) or (\bullet) to highlight the desired number.

Press the **Send** soft key to dial out the number.

Searching for Contacts

To search for a contact in the local directory:

1. Press the Directory soft key.

The IP phone enters the local directory directly as there is only Local Directory enabled in the directory by default.

	Local D	Directory	
1. All Conta	cts		
Back	AddGroup	Search	Enter

If Local Directory is removed from the directory (refer to Directory on page 61), press

Menu->Directory->Local Directory to enter the local directory.

- 2. Press the Search soft key.
- **3.** Enter a few continuous characters of the contact name or continuous numbers of the contact number (office, mobile or other number) using the keypad.



The contacts whose name or phone number matches the characters entered will appear on the LCD screen. You can dial from the result list.

Importing/Exporting Contact Lists

You can manage your phone's local directory via phone user interface or web user interface. But you can only import or export the contact list via web user interface.

To import an XML contact list file via web user interface:

- 1. Click on Directory->Local Directory.
- **2.** Click **Browse** to locate a contact list file (the file format must be *.xml) from your local system.

3. Click Import XML to import the contact list.

e alink 1466	Status	Account	Network	DSSKey Feat	tures Settings	Eng Directory	Log 0 ish(English) Security
Local Directory	Index	Name	Office Number M	Iobile Number Other Number	All Contacts 👻 🔲	NOTE	
	1	Зоу	<u>1000</u>	<u>1000</u> <u>1000</u>	All Contacts		
Remote Phone Book	2					Local Director The built-in pho	
BOOK	3					store the names	and phone
Phone Call Info	4					numbers of you	r contacts.
	5					You can add ne	
LDAP	7					contacts, edit, o for a contact, o	
Multicast IP	8					contact number directory.	from the local
	9						
Setting	10					You can import of contact list.	or export the
	Directory (Prev Next	Hang Up	Delete All Delete	Move To All Contar 🗸	You can cli more guides.	ck here to ge
	Office Number Mobile Number Other Number						
	Mobile Number	Auto	Ŧ	Delet	e Photo		
	Mobile Number Other Number Ring Tone				e Photo selected.		
	Mobile Number Other Number Ring Tone Group	All Co	ntacts 👻				
	Mobile Number Other Number Ring Tone Group Account	All Co Auto	ntacts 🗸	Browse No file Upload Photo	selected.		
	Mobile Number Other Number Ring Tone Group Account Photo	All Co Auto defau	ntacts • • Ilt_contact_ir •	Browse No file Upload Photo Import Local Directory	selected.		
	Mobile Number Other Number Ring Tone Group Account Photo Add	All Co Auto defau	ntacts 🗸	Browse No file Upload Photo Import Local Directory Browse No file selectory	selected. File ? ted.		
	Mobile Number Other Number Ring Tone Group Account Photo	All Co Auto defau	ntacts • • Ilt_contact_ir •	Browse No file Upload Photo Import Local Directory Browse No file selec Import XML Export	selected. File ? XML		
	Mobile Number Other Number Ring Tone Group Account Photo Add	All Co Auto defau	ntacts • • Ilt_contact_ir •	Browse No file Upload Photo Import Local Directory Browse No file selec Import XML Export Browse No file selec	selected.		
	Mobile Number Other Number Ring Tone Group Account Photo Add Group Setting	All Co Auto defau	ntacts • • Ilt_contact_ir •	Browse No file Upload Photo Import Local Directory Browse No file selec Import XML Export	selected.		

The web user interface prompts "The original contact will be covered, continue?".

4. Click **OK** to complete importing the contact list.

To import a CSV contact list file via web user interface:

- 1. Click on Directory->Local Directory.
- 2. Click **Browse** to locate a contact list file (the file format must be *.csv) from your local system.
- **3.** (Optional.) Check the **Show Title** checkbox.

It will prevent importing the title of the contact information which is located in the first line of the CSV file.

- 4. Click Import CSV to import the contact list.
- 5. (Optional.) Mark the **On** radio box in the **Delete Old Contacts** field.

It will delete all existing contacts while importing the contact list.

6. (Optional.) Select the contact information you want to import into the local directory from the pull-down list of **Index**.

Yealink 1466		_	_	_		-		-			Eng	Log Out
	Status	Acco	unt Netwo	ork	DSSKey		Feature	5	Settings		Directory	Security
Preview	Delete Old	Contacts 🔍 Or	n © Off								NOTE	
		play Name 👻	Office Number -	Ignore mobile_		Ignore other	r_number	Ignore	+ line	Iç	contacts-previe	ew-note
	2	Ad	1020						-1		You can cl more guides.	ick here to get
	3	jim sunmy	1003 5002						-1		more guides.	
	5	Tom	1004						-1			
	•	"	II							Þ		
				Impo	rt							

At least one item should be selected to be imported into the local directory.

7. Click **Import** to complete importing the contact list.

To export a contact list via web user interface:

- 1. Click on Directory->Local Directory.
- 2. Click Export XML (or Export CSV).
- 3. Click Save to save the contact list to your local system.

Note Importing/exporting contact lists is available via web user interface only.

Blacklist

The built-in phone directory can store names and phone numbers for a blacklist. You can store up to 30 contacts, add, edit, delete or search for a contact in the blacklist directory, and even call a contact from the blacklist directory. Incoming calls from blacklist directory contacts will be rejected automatically.

To add a contact to the blacklist directory manually:

- 1. Press Menu->Directory->Blacklist.
- 2. Press the Add soft key.

3. Enter the name and the office, mobile or other numbers in the corresponding fields.



4. Press () or () , or the **Switch** soft key to select the desired account from the **Account** field.

If **Auto** is selected, the phone will use the default account when placing calls to the contact from the blacklist directory.

5. Press the Save soft key to accept the change or the Back soft key to cancel.

To add a contact to the blacklist directory from the local directory:

1. Press Directory soft key.

The IP phone enters the local directory directly as there is only Local Directory enabled in the directory by default.

Local Directory						
L. All Conta	cts					

If Local Directory is removed from the directory (refer to Directory on page 61), press **Menu->Directory->Local Directory** to enter the local directory.

2. Select the desired contact group and then press the Enter soft key.

If the contact is not in any group, select All Contacts and then press the Enter soft key.

- **3.** Press (\bullet) or (\bullet) to highlight the desired contact.
- 4. Press the Option soft key, and then select Add to Blacklist from the prompt list.

The LCD screen prompts the following warning:



5. Press the **OK** soft key to accept the change.

For operating instructions on editing, deleting, placing calls to and/or searching for contacts in the blacklist directory, refer to the operating instructions of Editing Contacts on page 68, Deleting Contacts on page 68, Placing Calls to Contacts on page 71 and/or Searching for Contacts on page 72.

Remote Phone Book

You can add new contacts to the local directory, search for a contact, or simply dial a contact number from the remote phone book.

You can configure your new phone to access up to 5 remote phone books. For the access URL of the remote phone book, contact your system administrator.

For operating instructions on placing calls to and/or searching for contacts in the remote phone book, refer to the operating instructions of Placing Calls to Contacts on page 71 and/or Searching for Contacts on page 72.

Configuring an Access URL

To configure an access URL for a remote phone book via web user interface:

- 1. Click on Directory->Remote Phone Book.
- 2. Enter the access URL in the **Remote URL** field.

3. Enter the name in the **Display Name** field.

Yealink 1465	Status Account Network DS	Log (Englar(Englar) SKey Features Settings Directory Security	•
Local Directory	Index Remote URL	Display Name NOTE	
Remote Phone	1 http://192.168.0.10/Department.xml	Group One Remote Phone Book	
Book	2	It is a centrally maintained	
Phone Call Info	3	phone book, stored on the remote server.	
	4	Users only need the access U	
LDAP	5	of the remote phone book. T IP phone can establish a	ne
Multicast IP		connection with the remote server and download the pho	ne
Setting	Incoming/Outgoing Call Lookup	Enabled	'n
	Update Time Interval(Seconds)	21600 (?)	
	Confirm	Cancel	

4. Click **Confirm** to accept the change.

Note An access URL for a remote phone book is configurable via web user interface only.

Accessing the Remote Phone Book

To access your remote phone book via phone user interface:

1. Press Menu->Directory->Remote Phone Book.

If Remote Phone Book is added to the directory (refer to Directory on page 61), press **Directory->Remote Phone Book** to enter the remote phone book.

2. Select the desired remote group, and then press the Enter soft key.

The phone then connects to the remote phone book and proceeds to load it. The contacts in the remote phone book are displayed on the LCD screen.

Group o	ne		1/1 🕨		
🔔 Test1		23000			
🔔 Test2		303			
🔔 Test3		6650			
Back	Search	Option	Send		

3. Press the **Back** soft key to back to the previous screen.

Incoming/Outgoing Call Lookup

You can enable the phone to present the caller/callee identity stored in the remote phone book when receiving/placing a call.

To configure incoming/outgoing call lookup and update time interval via web user interface:

- 1. Click on Directory->Remote Phone Book.
- 2. Select Enabled from the pull-down list of Incoming/Outgoing Call Lookup.
- Enter the desired refresh period in the Update Time Interval(Seconds) field. The default value is 21600.

Yealink 1465	Status Acc	ount Network DS	SSKey Features Settings	Log Out English(English) • 5 Directory Security
Local Directory	Index	Remote URL	Display Name	NOTE
Remote Phone	1 http://192.	168.0.10/Department.xml	Group One	Remote Phone Book
Book	2			It is a centrally maintained phone book, stored on the
Phone Call Info	3			remote server.
LDAP	4			Users only need the access URL of the remote phone book. The
LDAP	5			IP phone can establish a connection with the remote
Multicast IP				server and download the phone
Setting	Incoming/Ou	itgoing Call Lookup	Enabled 👻 💡	book, and then display the remote phone book entries on
	Update Time	Interval(Seconds)	21600	the phone user interface.
		Confirm	Cancel	

4. Click **Confirm** to accept the change.

Call History Management

The SIP-T46G IP phone maintains call history lists of Missed Calls, Placed Calls, Received Calls and Forwarded Calls. Each call history list supports up to 100 entries. You can view call history, place a call, add a contact or delete an entry from the call history list.

History record feature is enabled by default. If you don't want to save the call history, you can disable the feature.

Viewing History Records

To view call history:

1. Press the **History** soft key.

The LCD screen displays all call records.

- 2. Press (•) or (•) to switch among All Calls, Missed Calls, Placed Calls, Received Calls and Forwarded Calls.
- **3.** Press (\bullet) or (\bullet) to select the desired entry.
- 4. Press the Option soft key, and then select Detail from the prompt list.

The detailed information of the entry appears on the LCD screen.

Placing a Call from History Records

To place a call from the call history list:

- 1. Press the **History** soft key.
- 2. Press () or () to switch among All Calls, Missed Calls, Placed Calls, Received Calls and Forwarded Calls.
- **3.** Press \frown or \frown to select the desired entry.
- 4. Press the Send soft key.

Adding a Contact to the Local Directory/Blacklist

To add a contact to the local directory (or blacklist directory) from the call history list:

- 1. Press the History soft key.
- 2. Press (•) or (•) to switch among All Calls, Missed Calls, Placed Calls, Received Calls and Forwarded Calls.
- **3.** Press (\bullet) or (\bullet) to select the desired entry.
- 4. Press the **Option** soft key, and then select **Add to Contact** (or **Add to Blacklist**) from the prompt list.
- 5. Enter the desired values in the corresponding fields.
- 6. Press the **Save** soft key.

For more information on local directory and/or blacklist, refer to Local Directory on page 62 and/or Blacklist on page 75.

Deleting History Records

To delete an entry from the call history list:

- 1. Press the **History** soft key.
- 2. Press (•) or (•) to switch among All Calls, Missed Calls, Placed Calls, Received Calls and Forwarded Calls.
- **3.** Press (\bullet) or (\bullet) to select the desired entry.
- 4. Press the **Delete** soft key.

To delete all entries from the call history list:

- 1. Press the **History** soft key.
- 2. Press or to switch among All Calls, Missed Calls, Placed Calls, Received Calls and Forwarded Calls.
- 3. Press the Option soft key, and then select Delete All from the prompt list.
- 4. Press the OK soft key.

The LCD screen prompts "Delete all the call records?".



5. Press the OK soft key to confirm the deletion or the Cancel soft key to cancel.

Disabling History Record

To disable history record via phone user interface:

- 1. Press Menu->Features->Others->General.
- **2.** Press (•) or (•), or the **Switch** soft key to select **Disabled** from the **History Record** field.

General					
1. Key as Send:	#	$\triangleleft \triangleright$			
2. History Record:	Disabled	$\triangleleft \triangleright$			

3. Press the Save soft key to accept the change or the Back soft key to cancel.

Search Source List in Dialing

You can search for a contact from the desired lists when the phone is on the dialing screen. The lists can be Local Directory, History, Remote Phone Book and LDAP.

To configure search source list in dialing via web user interface:

- 1. Click on Directory->Setting.
- 2. In the **Search Source List In Dialing** block, select the desired list from the **Disabled** column and click →.

The selected list appears in the **Enabled** column.

3. Repeat the step 2 to add more lists to the **Enabled** column.

- 4. To remove a list from the **Enabled** column, select the desired list and then click _____.
- 5. To adjust the display order of the enabled list, select the desired list, and click f or 1.

The LCD screen will display search results in the adjusted order.

Yealink 1466							Eng	Log Out glish(English) 🚽
	Status	Account	Network	DSSKey	Features	Settings	Directory	Security
Local Directory Remote Phone Book Phone Call Info LDAP Multicast IP Setting		ctory Disabled History Remote PI LDAP Ch Source List In Disabled LDAP	-	Enabled Local Directory Enabled Local Directory History Remote Phone Bo			the search sou the entered str results on the p Recent Call I It allows users placed calls list is on the pre-di	d lists. the in Dialing phone to earch entries from rce list based on ing, and display ore-dialing screen. In Dialing to view the when the phone aling screen. lick here to get
		Recent Cal	In Dialing Enabl	ed	↓ ↓ Cancel			

6. Click **Confirm** to accept the change.

Search source list in dialing is configurable via web user interface only.

To search for an entry in the enabled search source lists:

- 1. Pick up the handset, press the Speakerphone key or press the line key.
- **2.** Enter a few continuous characters of the entry's name or continuous numbers of the entry's phone number (office, mobile or other number) using the keypad.

The entries in the enabled search source lists whose name or phone number matches the characters entered will appear on the LCD screen. You can press \bigcirc or \bigcirc to scroll to the desired entry and then place a call to the entry.

♦) 1006			
% 1006	1		
	A	123	
	A	234	
	1001		
	101		
Send	123	Delete	Cancel

Note

System Customizations

Headset Use

If you want to use a wired headset, physically connect your headset to the phone and activate the headset mode for use. For more information on physically connecting a headset, refer to Phone Installation on page 20.

If you want to use a wireless headset, insert the Bluetooth USB dongle into the USB port at the back of the IP phone. For more information on using a Bluetooth headset, refer to Bluetooth on page 85.

This section provides an introduction to wired headset use.

Note

If both a Bluetooth headset and a wired headset are connected, only the Bluetooth headset can be used.

To use headset, make sure the headset mode is enabled. Contact your system administrator for more information.

Headset Mode Activation/Deactivation

To activate the headset mode:

1. Press (on the phone.

The HEADSET key LED illuminates solid green and the headset icon appears on the idle screen.

Press the line key or the **Answer** soft key to answer an incoming call. The call will be connected to your headset automatically.

Enter the desired number and then press the **Send** soft key, the phone will then place a call using the headset automatically. For more information on using the headset to place a call, refer to Placing Calls on page 117.

To deactivate the headset mode:

1. Press (**O**) again on the phone.

The HEADSET key LED goes out and the headset icon () disappears from the idle screen.

Headset Prior

You can use headset in priority when headset prior feature is enabled. This feature is especially useful for permanent or full-time headset users.

To enable headset prior via web user interface:

- 1. Click on Features->General Information.
- 2. Select Enabled from the pull-down list of Headset Prior.

ealink 1466							Log O English(English)
	Status	Account	Network	DSSKey	Features	Settings	Directory Security
Forward&DND	G	eneral Informatio	n				NOTE
General		Call Waiting		Enabled	•		Call Waiting
Information		Call Waiting On Co	de				It allows IP phones to receive new incoming call when there already an active call.
Audio				:			Auto Redial It allows IP phones to
Transfer		Allow Mute		Enabled	•		automatically redial a busy number after the first attempt
Call Pickup		Dual Headset		Enabled	•		Key As Send Assigns "#" or "*" as the send key.
Remote Control		Auto Answer Delay		1 Enabled			Hotline
Phone Lock		Headset Prior		Enabled	•		IP phone will automatically dial out the hotline number when lifting the handset, pressing th
ACD		DTMF Replace Tra	ı	Disabled	•		speakerphone key or the line key.
SMS		Accept SIP Trust S	Server Only	Disabled	•		Call Completion It allows users to monitor the
Action URL		Display Method on	Dialing	User Name	•		busy party and establish a call when the busy party becomes
Power LED		Auto Linekeys		Disabled	•		available to receive a call.
Notification Popups		Confir	n		Cancel		

3. Click **Confirm** to accept the change.

To use headset prior, you should activate the headset mode in advance:

- 1. Physically connect the headset.
- 2. Press (Q) to activate the headset mode.

Note If headset prior is enabled, the headset mode will not be deactivated until you press the **HEADSET** key again.

If headset prior is disabled, the headset mode can be deactivated by pressing the Speakerphone key or the **HEADSET** key.

Headset prior is configurable via web user interface only.

Dual Headset

You can use two headsets when dual headset feature is enabled. To use this feature, you must physically connect headsets to the headset jack and handset jack respectively. Once the phone connects to a call, the headset connected to the headset jack will have full-duplex capabilities, while the one connected to the handset jack will only be able to listen. To enable dual headset via web user interface:

- 1. Click on Features->General Information.
- 2. Select Enabled from the pull-down list of Dual Headset.

ealink 1466							
	Status	Account	Network	DSSKey	Features	Settings	Directory Security
Forward&DND	G	eneral Informati	on				NOTE
General		Call Waiting		Enabled	•		Call Waiting
Information		Call Waiting On Co	de				It allows IP phones to receive new incoming call when there
Audio							already an active call.
				:			Auto Redial
Intercom							It allows IP phones to automatically redial a busy
Transfer		Allow Mute		Enabled	•		number after the first attempt
Call Pickup		Dual Headset		Enabled	•		Key As Send Assigns "#" or "*" as the send
Remote Control		Auto Answer Dela	ıy(1∼4s)	1			key.
Remote Control		Enable Auto Answ	erTone	Enabled	*		Hotline IP phone will automatically dial
Phone Lock		Headset Prior		Enabled	-		out the hotline number when lifting the handset, pressing th
ACD		DTMF Replace Tra	n	Disabled	•		speakerphone key or the line key.
SMS		Accept SIP Trust	Server Only	Disabled	•		Call Completion
		Display Method or	Dialing	User Name	•		It allows users to monitor the busy party and establish a call
Action URL		Auto Linekevs	-	Disabled	-		when the busy party becomes available to receive a call.
Power LED							available to receive a Call.

3. Click **Confirm** to accept the change.

Note Dual headset is configurable via web user interface only.

Bluetooth Headset

The SIP-T46G IP phone supports Bluetooth. Bluetooth enables low-bandwidth wireless connections within a range of 10 meters (32 feet). The best performance is in the 1 to 2 meter (3 to 6 feet) range.

You can activate/deactivate the Bluetooth mode on the phone, and then pair and connect the Bluetooth headset with your phone. You can pair up to 30 Bluetooth headsets with your phone. However, only one headset can be connected at a time. You can also disconnect and delete your Bluetooth headset from the phone.

Note Ensure that the Bluetooth USB dongle is properly connected to the USB port at the back of the IP phone. For more information, refer to Phone Installation on page 20.

Activating the Bluetooth Mode

When you insert Bluetooth USB dongle BT40 into the USB port at the back of the IP phone, the LCD screen prompts "Adapter has been added, scanning the bluetooth devices right now?". You can press the **OK** soft key to enter the Bluetooth screen. You can activate Bluetooth mode and then scan Bluetooth devices.



To activate the Bluetooth mode via phone user interface:

- 1. Press Menu->Basic->Bluetooth.
- **2.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select **On** from the **Bluetooth** field.

Bluetooth						
1. Bluetooth:	On					
Back	Switch	Save				

3. Press the Save soft key to accept the change.

The LCD screen displays the Bluetooth icon 🛞 .

Bluetooth mode is configurable via web user interface at the path Features->Bluetooth.

Pairing and Connecting the Bluetooth Headset

Before you pair a Bluetooth headset, check the indicator on the Bluetooth headset to make sure that the Bluetooth headset is discoverable. For more information, refer to the documentation from the Bluetooth headset manufacturer.

To pair and connect your Bluetooth headset to your IP phone:

- 1. Press Menu->Basic->Bluetooth.
- 2. Press the Scan soft key to search the Bluetooth devices.

The LCD screen prompts the following message:

	Scan Bluetooth Device	
1 Note		
	Scanning	
Cancel		

The phone proceeds to scan the local Bluetooth devices. When the phone completes scanning, it will display the available Bluetooth devices on the LCD screen.

Scan Bluetooth Device		
🕵 Jabra Style v1.2.0	50:C9:71:	EF:3F:1F
Back	Refresh	Connect

- 3. Select the desired Bluetooth headset and then press **Connect** soft key.
- 4. Enter the desired PIN in the Password field.

The default PIN is "0000".

Enter Password			
Password:		••••	
Back	123	Delete	ОК

5. Press the OK soft key.

The Bluetooth headset is then automatically connected to your IP phone.

The Bluetooth icon 🛞 appears on the LCD screen.

With the Bluetooth headset paired and connected, you can use the Bluetooth headset to place and answer calls. For more information, refer to the documentation from the Bluetooth headset manufacturer.

Viewing the Bluetooth Headset Information

To view the Bluetooth headset information:

1. Press Menu->Basic->Bluetooth->Paired Bluetooth Device.

The LCD screen displays a list of paired and connected Bluetooth headsets.

	Paired Bluet	ooth Device	
PLT_M70		48:C1:AC	:F6:AF:2F
Back	Scan	Option	Disconnect

- 2. Select the desired Bluetooth headset.
- **3.** Press the **Option** soft key, and then select **Detail** from the prompt list to view the Bluetooth headset information:
 - Device Name: the name of the Bluetooth headset.
 - **MAC**: the MAC address of the Bluetooth headset.

1. Device Name:	PLT_M70
2. MAC:	48:C1:AC:F6:AF:2F

Editing Device Information

To edit device information via phone user interface:

1. Press Menu->Basic->Bluetooth->Edit My Device Information.

The LCD screen displays the device name and MAC address. The MAC address cannot be edited.

2. Enter the desired name in the **Device Name** field.

 Edit My Device Information

 1. Device Name:
 Yealink T46G

 2. MAC:
 00:18:91:DD:3C:61

 Back
 abc
 Delete
 Save

The default device name is "Yealink T46G".

3. Press the Save soft key to accept the change or the Back soft key to cancel.

The preconfigured Bluetooth device name will display in scanning list of other device.

Disconnecting the Bluetooth Headset

You can disconnect the Bluetooth headset from your phone. After you disconnect the Bluetooth headset, it will still remain paired and be displayed in the **Paired Bluetooth Device** list. So you can easily connect it to your IP phone again.

To disconnect your Bluetooth headset from your phone:

- 1. Press Menu->Basic->Bluetooth->Paired Bluetooth Device.
- 2. Select the connected Bluetooth headset.
- 3. Press the Disconnect soft key.

The Bluetooth headset is disconnected. To connect it to your IP phone again, press the **Connect** soft key.

Deleting the Paired Bluetooth Headset

You can delete your Bluetooth headset from your phone. When you delete the Bluetooth headset, it disappears from the **Paired Bluetooth Device** list. To connect it to your IP phone again, you need to scan and connect it again.

To delete your Bluetooth headset from your phone:

- 1. Press Menu->Basic->Bluetooth->Paired Bluetooth Device.
- 2. Select the Bluetooth headset you want to delete.
- 3. Press the Option soft key, and then select Delete from the prompt list.

The LCD screen prompts the following warning "Delete selected item?".

- 4. Press the **OK** soft key to confirm the deletion or the **Cancel** soft key to cancel.
- **Note** You can also delete all headsets by pressing the **Option** soft key and then select **Delete All**. For more information, refer to the above steps.

If you cannot connect a Bluetooth headset to the phone successfully, we recommend that you delete the Bluetooth headset from the **Paired Bluetooth Device** list (if it exists) before pairing it.

Deactivating the Bluetooth Mode

To deactivate the Bluetooth mode via phone user interface:

- 1. Press Menu->Basic->Bluetooth.
- **2.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select **Off** from the **Bluetooth** field.

Bluetooth		
1. Bluetooth:	Off	$\langle \rangle$
Back	Switch	Save

3. Press the Save soft key to accept the change.

The Bluetooth icon 🔧 disappears from the phone.

The Bluetooth headset won't be connected anymore, but it will still be paired. When you activate the Bluetooth mode again, the paired Bluetooth headset will appear in the **Paired Bluetooth Devices** list.

Bluetooth mode is configurable via web user interface at the path Features->Bluetooth.

DSS Keys

There are three types of DSS keys: Line Keys, Programable Keys and Ext Keys. Details will be introduced in the following. The SIP-T46G IP phone supports 27 line keys and 13 programable keys.

Line Keys

You can assign predefined functionalities to line keys. You can also define a label for a line key feature which will appear on the LCD screen. Line keys allow you to quickly access features such as recall and voice mail. The line key LEDs will indicate the monitored status when the line keys are assigned with particular features, such as BLF. The default key type of line key 1-16 is Line. The default key type of line key 17-27 is N/A, which indicates that this line key provides no

functionality until configuration.

To assign functionality to a line key via phone user interface:

- 1. Press Menu->Features->DSS Keys.
- 2. Select the desired line key, and then press the Enter soft key.
- 3. Select the desired key type from the **Type** field.
- 4. (Optional.) Select the desired key event type from the Key Type field.
- 5. (Optional.) Select the desired line from the Account ID field.
- 6. (Optional.) Enter the string that will appear on the LCD screen in the Label field.
- 7. (Optional.) Enter the corresponding value in the Value field.
- 8. (Optional.) Enter the corresponding value in the Extension field.
- 9. Press the Save soft key to accept the change or the Back soft key to cancel.

Line key is configurable via web user interface at the path Dsskey->Line Key.

Note When the phone is idle, you can also long press the line key to configure it directly on the phone.

You can assign functionalities to 27 line keys. These 27 line keys are indicated on three pages. When you assign functionality to line keys located in line key 1-10, the labels of line keys are indicated on the first page. The following figure shows an example:

<u> 1006</u>		11	36 Fri, Sep 02
1006			
			1005
<u>&</u> 1003			
History	Directory	DND	Menu

When you assign functionality to line key located in line key 11-27, the label of the line key 10-18 are indicated on the second page, and the labels of the line key 19-27 are indicated on the third page. The line key in the right bottom is used to switch pages. The following figure shows an example:



The following table lists the page icons:

Icons	Description
2	Indicates the current page of line keys.
1 3	Indicates the non-current page of line keys.

Page Tips

You can use page tips feature to enable the page icon and page switch key LED to indicate different statuses. This feature is disabled by default.

Icons	Description
1	Fast flashing: the BLF monitored user receives an incoming call on the non-current page. Solid: there is a parked call on the non-current page.
1	Fast flashing: the line receives an incoming call on the non-current page.

The following table lists the page icons to indicate different statuses:

For more information on the page switch key LED status, refer to LED Instructions on page 9.

To configure the page icon to indicate status via web user interface:

- 1. Click on Dsskey->Line Key.
- 2. Select Enabled from the pull-down list of Enable Page Tips.

alink 1465	Status		ccount	Net	work	D	sskey	Features	Settings	Directory
Line Key 1-9	Enable Page	Tips En:	abled	-		Labe	Length Defau	ilt 🔻		NOTE
	Key	I	уре		Value		Label	Line	Extension	
Line Key 10-18	Line Key1	Line	-	Default		•	1006	Line 1	•	Line Keys Line keys allow you to quickly
Line Key 19-27	Line Key2	Line	•	Default		Ŧ		Line 2	•	 access features such as recal and voice mail.
Programable Key	Line Key3	Line	•	Default		Ŧ		Line 3	•	
Ext Key	Line Key4	Line	•	Default		•		Line 4	•	You can click here to ge
EXERCY	Line Key5	Line	•	Default		Ŧ		Line 5	•	more guides.
	Line Key6	Line	•	Default		•		Line 6	•	
	Line Key7	Line	•	Default		Ŧ		Line 7	•	
	Line Key8	Line	-	Default		Ŧ		Line 8	•	
	Line Key9	Line	•	Default		•		Line 9		-

3. Click **Confirm** to accept the change.

Label Length

You can specify labels for some key features, which will be displayed on the idle LCD screen. If the length of the label exceeds the maximum display length, the phone only displays the first few characters. You can configure the phone to display more characters via label length feature.

The following figure shows an example of label display when the label length feature is set to Extended:



The following figure shows an example of label display when the label length feature is set to Default:



To configure the label length via web user interface:

- 1. Click on Dsskey->Line Key.
- 2. Select Extended from the pull-down list of Label Length.

alink 1465	Status	Acc	count	Network	D	sskey	Features	Settings	Directory Security
Line Key 1-9	Enable Page	Tips Enab	led	•	Labe	Length Exte	nded 👻]	NOTE
	Key	Typ	be	Value		Label	Line	Extension	
Line Key 10-18	Line Key1	Line	•	Default	•	1006	Line 1	·	Line Keys Line keys allow you to quickly
Line Key 19-27	Line Key2	Line	•	Default	•		Line 2	•	 access features such as recal and voice mail.
Programable Key	Line Key3	Line	•	Default	•		Line 3	·	
Ext Key	Line Key4	Line	•	Default	•		Line 4	•	You can click here to get
LACINCY	Line Key5	Line	•	Default	•		Line 5	•	more guides.
	Line Key6	Line	•	Default	•		Line 6	•	
	Line Key7	Line	Ŧ	Default	•		Line 7	·	
	Line Key8	Line	Ŧ	Default	¥		Line 8	•	
	Line Key9	Line	•	Default	-		Line 9		

3. Click **Confirm** to accept the change.

Note Page tips and label length features are configurable via web user interface only.

Line Key Features

Line key features are explained in the following subchapters in detail:

- Line
- Speed Dial
- Voice Mail
- Direct Pickup
- Group Pickup
- Call Park

- DTMF
- Prefix
- Local Group
- XML Group
- XML Browser
- LDAP
- Conference
- Forward
- Transfer
- Hold
- DND
- SMS
- Group Listening
- Zero Touch
- URL
- Phone Lock
- Retrieve
- Directory

For the features not listed above, refer to Basic Call Features on page 117 and Advanced Phone Features on page 163. For more information, contact your system administrator.

Line

You can use this key feature to accept incoming calls, place active calls on hold or resume a held call. It performs in the same way as a hard line key.

Dependencies: Type (Line)

Account ID (the account this feature will be applied to) Label (key label displayed on the LCD screen) Value (Default/Lock)

Usage: When the phone receives an incoming call, the Line key LED flashes green:

- 1. Press the Line key to accept the incoming call.
- 2. Press the Line key to place a new call and the active call is placed on hold.
- 3. Press the Line key again to resume the held call.

If you select **Lock** from the **Value** field, the line keys located in the same position of the other page will be automatically assigned Line feature with **Lock** filled in the **Value** field. This feature is only applicable to line key 1-9. Then you can use this key on every page. If you select **Default** from the **Value** field, you can only use this key on the current page.

Speed Dial

You can use this key feature to speed up dialing the numbers frequently used or hard to remember.

Dependencies: Type (Speed Dial)

Account ID (the account this feature will be applied to) *Label* (key label displayed on the LCD screen) *Value* (the number you want to dial out)

Usage: Press the **Speed Dial** key to dial out the number specified in the **Value** field, using the account selected from the **Account ID** field.

Voice Mail

You can use this key feature to quickly connect voice mail. For more information, refer to Voice Mail on page 215.

Dependencies: Type (Key Event)

Key Type (Voice Mail) Account ID (the account this feature will be applied to) *Label* (key label displayed on the LCD screen) *Value* (the voice mail access code)

Usage: Press the **Voice Mail** key to dial out the voice mail access code. Then follow the voice prompt to listen to the voice mails.

Direct Pickup

You can use this key feature to answer someone else's incoming call on the phone.

Dependencies: Type (Key Event)

Key Type (DPickup) Account ID (the account this feature will be applied to) *Label* (key label displayed on the LCD screen) *Value* (the directed call pickup code followed by the target phone number)

Usage: Press the **Direct Pickup** key on your phone when the target phone number receives an incoming call. The call is then answered on your phone.

Group Pickup

You can use this key feature to answer incoming calls in a group that is associated with their own group.

Dependencies: Type (Key Event)

Key Type (GPickup) Account ID (the account this feature will be applied to) *Label* (key label displayed on the LCD screen) *Value* (the group call pickup code) **Usage:** Press the **Group Pickup** key on your phone when a phone number in the group receives an incoming call. The call is answered on your phone.

Call Park

You can use this key feature to park a call.

Dependencies: Type (Key Event)

Key Type (Call Park) Account ID (the account this feature will be applied to) *Label* (key label displayed on the LCD screen) *Value* (the call park code)

Usage: Press the Call Park key on your phone to park a call.

DTMF

You can use this key feature to send the specification of arbitrary key sequences via DTMF.

Dependencies: Type (Key Event)

Key Type (DTMF) Label (key label displayed on the LCD screen) Value (DTMF sequence)

Usage: Press the **DTMF** key during an active call to send the key sequence specified in the **Value** field.

Note DTMF sequence can only contain "0-9", "*", "#" and "A-E".

Prefix

You can use this key feature to add a specified prefix number before the dialed number.

Dependencies: Type (Key Event)

Key Type (Prefix) Label (key label displayed on the LCD screen) *Value* (the prefix number)

Usage: Press the **Prefix** key when the phone is idle, the phone will then enter the dialing screen and display the prefix number that you specified in the **Value** field. You can enter the remaining digits and then dial out.

Local Group

You can use this key feature to quickly access a contact group in the local directory. For more information, refer to Local Directory on page 62.

Dependencies: Type (Key Event)

Key Type (Local Group) Local Group (the contact group name you want to access) Label (key label displayed on the LCD screen)

Usage: Press the **Local Group** key to access the contact group specified in the **Local Group** field.

XML Group

You can use this key feature to quickly access a remote group in your remote phone book. You should configure remote phone book in advance. For more information, refer to Remote Phone Book on page 77.

Dependencies: Type (Key Event)

Key Type (XML Group) XML Group (the remote group name you want to access if remote phone book is configured)

Label (key label displayed on the LCD screen)

Usage: Press the XML Group key to access the remote group specified in the XML Group field.

XML Browser

You can use this key feature to quickly access an XML browser. The XML browser allows you to create custom services which meet your functional requirements on the server. You can customize practical applications, such as weather report, stock information, Google search, etc.

Dependencies: Type (Key Event)

Key Type (XML Browser) Label (key label displayed on the LCD screen) *Value* (the access URL for XML browser)

Usage: Press the XML Browser key to access the XML browser specified in the Value field.

LDAP

You can use this key feature to quickly access a LDAP search screen.

Dependencies: Type (Key Event)

Key Type (LDAP)

Label (key label displayed on the LCD screen)

Usage:

- 1. Press the LDAP key to access the LDAP search screen.
- **2.** Enter a few continuous characters of the contact name or continuous numbers of the contact number using the keypad.

The contacts whose name or phone number matches the characters entered will appear on the LCD screen.

Conference

You can use this key feature to set up a conference call. For more information, refer to Conference on page 147.

Dependencies: Type (Key Event)

Key Type (Conference)

Label (key label displayed on the LCD screen)

Value (the number you want to add to the conference)

Usage: Press the **Conference** key during an active call to set up a conference with the number specified in the **Value** field.

Note If the **Value** field is left blank, the **Conference** key performs the same as the **Conference** soft key during a call.

Forward

You can use this key feature to forward an incoming call to someone else. For more information, refer to Call Forward on page 135.

Dependencies: Type (Key Event)

Key Type (Forward) Label (key label displayed on the LCD screen) Value (the number you want to forward to)

Usage:

Press the Forward key to forward an incoming call to the number specified in the Value field.

Note If the **Value** field is left blank, the **Forward** key performs the same as the **Forward** soft key when receiving an incoming call.

Transfer

When there is an active call on the phone, you can use this key feature to handle the call differently depending on the transfer mode assigned to the DSS key.

Dependencies: Type (Key Event)

Key Type (Transfer)

Label (key label displayed on the LCD screen)

Value (the number you want to transfer to)

Usage:

- When the transfer mode on DSS key is **Blind Transfer**, press the **Transfer** key to complete the blind transfer to the number specified in the **Value** field.
- When the transfer mode on DSS key is Attended Transfer, press the Transfer key to dial out the number specified in the Value field, and then perform the attended or semi-attended transfer.

• When the transfer mode on DSS key is **New Call**, press the **Transfer** key to place a new call to the number specified in the **Value** field.

Note Transfer mode via DSS key is configurable via web user interface at the path Features->Transfer->Transfer Mode Via Dsskey.

If the **Value** field is left blank, the **Transfer** key performs the same as the **TRANSFER** key or the **Transfer** soft key during a call. For more information, refer to Call Transfer on page 143.

Hold

You can use this key feature to place an active call on hold or retrieve a held call.

Dependencies: Type (Key Event)

Key Type (Hold) Label (key label displayed on the LCD screen)

Usage:

- 1. Press the Hold key during an active call to place the call on hold.
- 2. Press the Hold key again to retrieve the held call.

DND

You can use this key feature to enable or disable DND. You can also use this key feature to access the custom DND screen. For more information, refer to Do Not Disturb (DND) on page 130.

Dependencies: Type (Key Event)

Key Type (DND)

Label (key label displayed on the LCD screen)

Usage:

When DND is in phone mode:

- 1. Press the **DND** key to enable DND.
- 2. Press the DND key again to disable DND.

When DND is in custom mode:

 Press the **DND** key to access the custom DND screen. You can enable or disable DND for one or all accounts.

Note When DND is activated, the **DND** key LED will illuminate solid green, and the incoming calls will be rejected automatically.

SMS

You can use this key feature to quickly access text message. For more information, refer to Short Message Service (SMS) on page 213.

Dependencies: Type (Key Event)

Key Type (SMS)

Label (key label displayed on the LCD screen)

Usage: Press the SMS key when the phone is idle to access text message.

Group Listening

You can use this key feature to activate the Speakerphone and Handset/Headset mode at the same time. It is suitable for the group conversations which have more than one person present at one end. You are able to speak and listen through the handset/headset, while the others nearby can only listen through the speaker.

Dependencies: Type (Key Event)

Key Type (Group Listening)

Label (key label displayed on the LCD screen)

Usage:

1. During a call, press the Group Listening key to activate the group listening mode.

You can then speak and listen through the handset/headset, while other people at your side can only listen through the speaker at the same time.

2. Press the Group Listening key again to deactivate the group listening mode.

Zero Touch

You can use this key feature to quickly configure auto provision and network parameters.

Dependencies: Type (Key Event)

Key Type (Zero Touch) Label (key label displayed on the LCD screen)

Usage:

- 1. Press the Zero Touch key to access the zero touch screen.
- 2. Press the OK soft key within a few seconds.
- 3. Configure the network parameters in the corresponding fields.
- 4. Press the Next soft key.
- 5. Configure the auto provision parameters in the corresponding fields.
- 6. Press the OK soft key.

The phone will reboot to update configurations.

URL

You can use this key feature to trigger the phone to send an HTTP GET request containing a specific URL.

Dependencies: Type (URL)

Label (key label displayed on the LCD screen) *URL* (the URL contained in the HTTP GET request) **Usage:** Press the **URL** key to trigger the phone to send an HTTP GET request containing the URL specified in the **URL** field.

Phone Lock

You can use this key feature to immediately lock your phone instead of long pressing $[\#_{mo}]$. For more information, refer to Phone Lock on page 53.

Dependencies: Type (Key Event)

Key Type (Phone Lock)

Label (key label displayed on the LCD screen)

Usage: When the phone lock feature is enabled, press the **Phone Lock** key to immediately lock your phone instead of long pressing $\#_{\text{tess}}$.

Retrieve

You can use this key feature to retrieve a parked call.

Dependencies: Type (Retrieve)

Account ID (the account this feature will be applied to) Label (key label displayed on the LCD screen) Value (the park retrieve code)

Usage: Press the Retrieve key on your phone to retrieve a parked call.

Directory

You can use this key feature to easily access frequently used lists. For more information, refer to Directory on page 61.

Dependencies: Type (Key Event)

Key Type (Directory)

Label (key label displayed on the LCD screen)

Usage: Press the Directory key to immediately access to frequently used lists.

Note The **Directory** key performs the same function as the **Directory** soft key when the phone is idle.

Programable Keys

You can customize the soft keys, navigation keys and function keys.

To customize programable keys via web user interface:

- 1. Click on Dsskey->Programable Key.
- 2. Customize specific features for these keys.

alink T46G							English
	Status	Account	Network	Dsskey	Features	Settings	Directory Security
Line Key 1-9	Кеу	Туре	Line	Value	Label	Extension	NOTE
	SoftKey 1	History -	Local History 👻				
Line Key 10-18	SoftKey 2	Directory -	N/A 🚽				 Programmable Keys Customizes the soft keys, navigation keys and function keys.
Line Key 19-27	SoftKey 3	DND 👻	N/A 🚽				
Programable Key	SoftKey 4	Menu 👻	N/A 🚽				
Ext Key	Up	History 👻	Local History 👻				You can click here to ge more guides.
LACKCY	Down	Directory 👻	N/A 🚽				more galacs.
	Left	Switch Account L 👻	N/A 🚽				
	Right	Switch Account [👻	N/A 🚽				
	ОК	Status 👻	N/A 👻				
	Cancel	N/A 👻	N/A 🚽				
	HOLD	N/A 👻	N/A 👻				
	Mute	N/A 👻	N/A 👻				
	Tran	Forward 👻	N/A -				

3. (Optional.) Enter the string that will appear on the LCD screen in the **Label** field.

Label is configurable only when customizing SoftKey (1-4).

4. Click **Confirm** to accept the change.

Note Programable keys are configurable via web user interface only.

The MUTE key cannot be customized when keep mute feature is enabled. For more information, contact your system administrator.

You can click Reset To Default to reset custom settings to defaults.

Then you can press the keys on the phone to perform the features you configured.

For example:

Switch Account Up

You can use this key feature to change the default account.

Dependencies: Type (Switch Account Up)

Usage: Press the **Switch Account Up** key to scroll up the account list to select the desired default account.

Switch Account Down

You can use this key feature to change the default account.

Dependencies: Type (Switch Account Down)

Usage: Press the Switch Account Down key to scroll down the account list to select the desired

default account.

Ext Keys

If EXP40 is connected to the phone, you can customize features for ext keys.

To customize ext keys via web user interface:

- 1. Click on Dsskey->Ext Key.
- 2. Customize specific features for these keys.

	Status	;	Account	t Network	Dsskey	Fea	tures Settings	Directory Security
ine Key 1-9	Expansion	Кеу	Expansion 1	•				NOTE
ine Key 10-18	Кеу		Туре	Value	Label	Line	Extension Key	Ext Keys
	Key1	N/A	•		N/A	·		Customizes the key feature of the expansion module.
ine Key 19-27		N/A	•		N/A	۰ ۲	Key2	che expansion module.
Programable Key	Key3	N/A	•		N/A			You can click here to ge
xt Key		N/A	•		N/A	·	Кеу4	more guides.
	Key5	N/A	•		N/A	F		
		N/A	×		N/A	 	Кеуб	
	Key7	N/A	•		N/A	A		
		N/A	•		N/A	а – Г	Кеу8	
	Key9	N/A	•		N/#	· · · · ·		
		N/A	•		N/#	A	Key10	
	Key11	N/A	•		N//	. → [
		N/A	•		N/A	· ·	Key12	

3. Click **Confirm** to accept the change.

For more information, refer to Yealink EXP40 User Guide.

Note

Ext keys are configurable via web user interface only.

Account Management

You can register one or multiple accounts on the SIP-T46G IP phone. You can also configure each line key to associate with an account or configure multiple line keys to associate with an account.

Account Registration

To register an account via phone user interface:

- 1. Press Menu->Advanced (default password: admin) ->Accounts.
- 2. Select the desired account and then press the Enter soft key.
- 3. Select Enabled from the Activation field.
- 4. Enter the desired value in Label, Display Name, Register Name, User Name, Password

and **SIP Server1/2** field respectively. Contact your system administrator for more information.

- 5. If you use the outbound proxy servers, do the following:
 - Press (•) or (•), or the Switch soft key to select Enabled from the Outbound Status field.
 - Enter the desired value in the Outbound Proxy1/2 and Fallback Interval field respectively. Contact your system administrator for more information.
- 6. Press the Save soft key to accept the change or the Back soft key to cancel.

You can repeat steps 2 to 6 to register more accounts.

The following figures demonstrate single or multiple accounts registered on the phone:

Single account:



Multiple accounts:

1006		13	:27 Fri, Sep 02
1006			
1007			
History	Directory	DND	Menu

To disable an account via phone user interface:

- 1. Press Menu->Advanced (default password: admin) ->Accounts.
- 2. Select the desired account and then press the Enter soft key.
- 3. Select **Disabled** from the **Activation** field.
- 4. Press the Save soft key to accept the change or the Back soft key to cancel.

Registering an account is configurable via web user interface at the path Account->Register.

Default Account

To configure the default account via phone user interface:

- 1. Press Menu->Features->Others->Default Account.
- **2.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select the desired account.

Def	ault Account	
. Default Account:	1046@10.2.1.48	$\triangleleft \triangleright$

3. Press the **Save** soft key to accept the change or the **Back** soft key to cancel.

Default account can be set by pressing \bigcirc or \bigcirc on the idle screen. It has priority when placing a call.

Note The phone's default account cannot be changed after reboot.

Multiple Line Keys per Account

You can configure multiple line keys to associate with an account. This enhances call visualization and simplifies call handling.

	Status	Acco	ount	Network	D	sskey	Features	Settings	Directory Security
Line Key 1-9	Enable Page	Tips Disable	ed	•	Labe	Length Defa	ult 👻		NOTE
	Кеу	Тур	e	Value		Label	Line	Extension	
Line Key 10-18	Line Key1	Line	-	Default	•	1006	Line 1	,	Line Keys Line keys allow you to quickly
Line Key 19-27	Line Key2	Line	•	Default	•	1006	Line 1	-	access features such as recall and voice mail.
Programable Key	Line Key3	Line	-	Default	•		Line 3	-	
Ext Key	Line Key4	Line	-	Default	•		Line 4	-	You can click here to ge more guides.
LAC NOT	Line Key5	Line	•	Default	•		Line 5	-	more guides.
	Line Key6	Line	•	Default	•		Line 6	-	
	Line Key7	Line	•	Default	•		Line 7	-	
	Line Key8	Line	•	Default	•		Line 8	-	
	Line Key9	Line	•	Default	-		Line 9	-	

1006		13	:29 Fri, Sep 02
1006			
2 1006			
History	Directory	DND	Menu

If this is the case, the LCD screen will resemble the following figure:

Incoming calls to this account will be distributed evenly among the available line keys. Outgoing calls will be distributed similarly.

Your phone can be configured to have a combination of accounts with a single line key and accounts with multiple line keys.

Dial Plan

Dial plan is a string of characters that governs the way your SIP-T46G IP phone processes the inputs received from your phone keypad.

The SIP-T46G IP phone supports the following dial plan features:

- Replace Rule
- Dial Now
- Area Code
- Block Out

The basic expression syntax you need to know:

	The dot "." can be used as a placeholder or multiple placeholders for any character. Example:
•	"12." would match "12 3 ", "12 34 ", "12 345 ", "12 abc ", etc.
v	An "x" can be used as a placeholder for any character. Example:
X	"12x" would match "12 1 ", "12 2 ", "12 3 ", "12 a ", etc.
	Numeric ranges are allowed within the brackets: Digit "-" Digit. Example:
	"[5-7]" would match the number" 5 ", " 6 "or " 7 ".
	The square brackets "[]" can be used as a placeholder for a single character
[]	which matches any of a set of characters. Example:
	"91[5-7]1234" would match "91 5 1234", "91 6 1234", "91 7 1234".
	The parentheses "()" can be used to group together patterns, for instance, to
0	logically combine two or more patterns. Example:
	"([1-9])([2-7])3" would match " 92 3", " 15 3", " 77 3", etc.

The "\$" followed by the sequence number of a parenthesis means the
characters placed in the parenthesis. The sequence number stands for the
corresponding parenthesis. Example:
\$ A replace rule configuration, Prefix: "001(xxx)45(xx)", Replace: "9001\$145\$2".
When you dial out "0012354599" on your phone, the IP phone will replace the
number with "9001 235 45 99 ". "\$1" means 3 digits in the first parenthesis, that
is, "235". "\$2" means 2 digits in the second parenthesis, that is, "99".

Note The IP phone supports a new dial plan mechanism – digit map. You can use one or more matching rules in one digit map string. If the new dial plan mechanism is enabled, old dial plan will be ignored. For more information, contact your system administrator.

Replace Rule

You can configure one or more replace rules (up to 100) to remove the specified string and replace it with another string. You can configure a pattern with wildcards (refer to the expression syntax in the table above), so that any string that matches the pattern will be replaced. This feature is convenient for you to dial out a long number. For example, a replace rule is configured as "Prefix: 1" and "Replace: 1234567", when you try to dial out the number "1234567", you just need to enter "1" on the phone and then press the Send soft key.

To add a replace rule via web user interface:

- 1. Click on Settings->Dial Plan->Replace Rule.
- 2. Enter the string (e.g., 1) in the **Prefix** field.
- 3. Enter the string (e.g., 1234567) in the Replace field.
- 4. Enter the desired line ID in the **Account** field or leave it blank.

	Status	Account	Network DS	SKey Features	Settings	Directory Security
Preference	Replace F	Rule Dial Now	Area Code Block 0	ut		NOTE
Time & Date	Index	Prefix	Replace	Account		Replace Rule:An alternative
	1					string that replaces the entere
Call Display	2					Dial-now:Automatically dial or
Upgrade	3					the entered numbers. Area Code:Automatically add
1.5	4					the area code before the
Auto Provision	5					numbers when dialing. Block Out:It prevents users
Configuration	6					from dialing out specific
	7					"."represents any string.
Dial Plan	8					"x"represents any character.
Voice	9					"-"match a range of characte within the brackets.
	-					",":a separator within the
Ring	10					bracket. "[]":a character matches any
Tones						character sets. "O":combines two or more
						patterns.

5. Click Add to add the replace rule.

When you enter the number "1" using the keypad and then press the Send soft key, the

phone will dial out "1234567" instead.

Note The valid values for the **Account** field can be one or more digits among 1-16. Every two digits must be separated by a comma. For example, when you enter the value "1, 2" in the **Account** field, this replace rule will apply to account 1 and account 2.

If you leave the Account field blank or enter 0, the replace rule will apply to all accounts.

To edit a replace rule via web user interface:

- 1. Click on Settings->Dial Plan->Replace Rule.
- 2. Select the desired replace rule by checking the checkbox.
- 3. Edit the values in the Prefix and Replace fields.
- 4. Enter the desired line ID in the Account field or leave it blank.
- 5. Click Edit to accept the change.

To delete one or more replace rules via web user interface:

- 1. Click on Settings->Dial Plan->Replace Rule.
- 2. Select the one or more replace rules by checking the checkbox(es).
- **3.** Click **Del** to delete the replace rule(s).

Replace rule is configurable via web user interface only.

Dial Now

Note

You can configure one or more dial now rules (up to 100) on your phone. When the dialed number matches the dial now string, the number will be dialed out automatically. For example, a dial now rule is configured as "1xxx", any entered four-digit string beginning with 1 will then be dialed out automatically on the phone.

To add a dial now rule via web user interface:

- 1. Click on Settings->Dial Plan->Dial Now.
- 2. Enter the desired value (e.g., 1xxx) in the Rule field.
- 3. Enter the desired line ID in the **Account** field or leave it blank.

For more information on the valid values for the **Account** field, refer to Replace Rule on page 108.

	Status	Account	Network	DSSKey	Features	Settings	Directory Security
Preference	Replace R	ule Dial Now	Area Code	Block Out			NOTE
lime & Date	Index	Dial-now Rul	•		Account		Replace Rule:An alternative
	1						string that replaces the enter
Call Display	2						numbers. Dial-now:Automatically dial of
Upgrade	3						the entered numbers. Area Code:Automatically ad
opgrade	4						the area code before the
Auto Provision							numbers when dialing. Block Out: It prevents users
0	5						from dialing out specific
Configuration	6						numbers.
Dial Plan	7						"."represents any string.
	8						"x"represents any character "-"match a range of character
Voice	9						within the brackets.
Ring	10						",":a separator within the bracket.
							"[]":a character matches any
Tones							character sets. "Ω":combines two or more
Softkey Layout							patterns. "\$":followed by the sequence

4. Click Add to add the dial now rule.

When you enter the number "1234" using the keypad, the phone will dial out "1234" automatically without the pressing of any key.

You can also edit or delete the dial now rule, refer to Replace Rule on page 108 for more information.

Dial-now rule is configurable via web user interface only.

Time Out for Dial Now Rule

You can configure the delay time for dial now rules. That is, you can configure your phone to automatically dial out the phone number which matches a dial now rule, after the designated delay time.

Note

To configure the delay time for dial now rule via web user interface:

- 1. Click on Features->General Information.
- 2. Enter the time between 0 and 14 (seconds) in the **Time Out for Dial Now Rule** field.

The default value is "1".

			Log Out
Yealink 1466			
	Status Account Network	DSSKey Features	Settings Directory Security
Forward&DND	General Information		NOTE
General	Call Waiting	Enabled 🔻	Call Waiting
Information	Call Waiting On Code		It allows IP phones to receive a new incoming call when there is
Audio	Call Waiting Off Code		already an active call.
	Auto Redial	Disabled 🔹	Auto Redial It allows IP phones to
Intercom	Auto Redial Interval (1~300s)	10	automatically redial a busy number after the first attempt.
Transfer	Auto Redial Times (1~300)	10	Key As Send
Call Pickup	Key As Send	#	Assigns "#" or "*" as the send key.
Remote Control	Reserve # in User Name	Enabled 🔻	Hotline IP phone will automatically dial
Phone Lock	Hotline Number		out the hotline number when lifting the handset, pressing the
Thomas Local	Hotline Delay(0~10s)	4	speakerphone key or the line key.
ACD	Busy Tone Delay (Seconds)	0	Call Completion
SMS	Return Code When Refuse	486 (Busy Here)	It allows users to monitor the busy party and establish a call
Action URL	Return Code When DND	480 (Temporarily Unava 🔻	when the busy party becomes available to receive a call.
Power LED	Call Completion	Disabled 🔹	You can click here to get
	Feature Key Synchronization	Disabled 🔻	more guides.
Notification Popups	Time Out for Dial Now Rule	1	
	RFC 2543 Hold	Disabled 🔻	

3. Click **Confirm** to accept the change.

Time out for dial now rule is configurable via web user interface only.

Area Code

Note

Area codes are also known as Numbering Plan Areas (NPAs). They usually indicate geographical areas in a country. This feature is necessary when dialing a phone number outside the code area. For example, an area code is configured as "Code: 0592, Min Length: 1, Max Length: 15". When you dial out the number "56789" (the length of the number is between 1 and 15), the phone will add the area code and dial out the number "059256789". You can only configure one area code rule on your phone.

To configure the area code via web user interface:

- 1. Click on Settings->Dial Plan->Area Code.
- 2. Enter the desired values in the Code, Min Length (1-15) and Max Length (1-15) fields.
- 3. Enter the desired line ID in the Account field or leave it blank.

For more information on the valid values for the **Account** field, refer to Replace Rule on page 108.

Yealink 146G	Status Account	Network	DSSKey	Features	Settings	Log Out English(English) - Directory Security
Preference	Replace Rule Dial Now	Area Code Blo	ck Out			NOTE
Time & Date		Code	0592	7		Replace Rule:An alternative string that replaces the entered
Call Display		Min Length (1-15)	1	Í.		numbers. Dial-now:Automatically dial out
Upgrade		Max Length (1-15) Account	15	_		the entered numbers. Area Code:Automatically add
Auto Provision	Con	firm	ſ	Cancel		the area code before the numbers when dialing. Block Out: It prevents users
Configuration			l			from dialing out specific numbers.
Dial Plan						"."represents any string. "x"represents any character.

4. Click **Confirm** to accept the change.

Note

The default value of minimum and maximum length is 1 and 15 respectively.

Area code is configurable via web user interface only.

Block Out

You can block some specific numbers (up to 10) from being dialed on your phone. When you dial a block out number on your phone, the dialing will fail and the LCD screen will prompt "Forbidden Number".

To add a block out number via web user interface:

- 1. Click on Settings->Dial Plan->Block Out.
- 2. Enter the desired value in the **BlockOut NumberX** field.
- 3. Enter the desired line ID in the Account field or leave it blank.

For more information on the valid values for the **Account** field, refer to Replace Rule on page 108.

ealink 146G	Status	ount Network	DSSKey Features	Settings	Log C English(English) Directory Security
Preference	Replace Rule Di	al Now Area Code	Block Out		NOTE
Time & Date	BlockOut Number 1	4321	Account		Replace Rule:An alternative string that replaces the entered
Call Display	BlockOut Number 2		Account		numbers.
	BlockOut Number 3		Account		Dial-now:Automatically dial ou the entered numbers.
Upgrade	BlockOut Number 4		Account		Area Code:Automatically add
Auto Provision	BlockOut Number 5		Account		the area code before the numbers when dialing.
Auto Provision	BlockOut Number6		Account		Block Out: It prevents users
Configuration	BlockOut Number 7		Account		from dialing out specific numbers.
3	BlockOut Number8		Account		
Dial Plan	BlockOut Number9		Account		"."represents any string. "x"represents any character.
Voice	BlockOut Number 10		Account		 "match a range of character within the brackets.
Ring		Confirm	Cancel		",":a separator within the bracket.

4. Click **Confirm** to add the block out number.

Note Block out number is configurable via web user interface only.

Emergency Number

Public telephone networks in countries around the world have a single emergency telephone number (emergency services number), that allows a caller to contact local emergency services for assistance when necessary. The emergency telephone number may differ from country to country. It is typically a three-digit number so that it can be easily remembered and dialed quickly. Some countries have a different emergency number for each of the different emergency services.

You can specify the emergency telephone numbers on the IP phone for contacting the emergency services in an emergency situation. You can dial these numbers when the phone is locked. For more information on phone lock, refer to Phone Lock on page 53.

Note Contact your local phone service provider for available emergency numbers in your area. The IP phone also supports the emergency dialplan. For more information, contact your system administrator.

To specify emergency numbers via web user interface:

- 1. Click on Features->Phone Lock.
- 2. Enter the emergency number in the Emergency field.

For multiple emergency numbers, enter a comma between every two numbers. The default emergency numbers are 112, 911 and 110.

Yealink 1466	Status Account Network	DSSKey Features	Log Out English(English) • Settings Directory Security
Forward&DND	Phone Lock Enable	Disabled 👻 🕜	NOTE
General Information	Phone Lock Type Phone Unlock PIN(0~15 Digit)	All Keys • ?	Phone Lock It is used to lock the IP phone
Audio	Phone Lock Time Out(0~3600s)	0	to prevent it from unauthorized use. Once the IP phone is locked, a user must enter the password to unlock it.
Intercom Transfer	Confirm	Cancel	IP phones offer three types of phone lock: Menu Key, Function Keys and All Keys.
Call Pickup Remote Control			The IP phone will not be locked immediately after the phone lock type is configured.
Phone Lock			You can click here to get more guides.

3. Click **Confirm** to accept the change.

Note

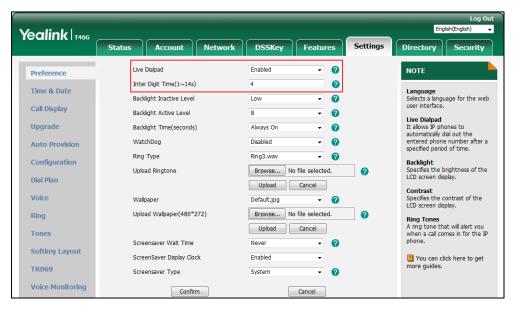
Emergency number is configurable via web user interface only.

Live Dialpad

You can enable live dialpad feature on the SIP-T46G IP phone, which enables the IP phone to automatically dial out a phone number without pressing the send key. You can also configure a delay, and then the phone will dial out the phone number automatically after the designated period of time.

To enable the live dialpad via web user interface:

- 1. Click on Settings->Preference.
- 2. Select Enabled from the pull-down list of Live Dialpad.
- Enter the desired delay time in the Inter Digit Time(1~14s) field. The default delay time is 4s.



4. Click **Confirm** to accept the change.

Note Live dialpad is configurable via web user interface only.

Hotline

You can dial a hotline number immediately upon lifting the handset, pressing the Speakerphone key or the line key. You can also configure a delay, and then the phone will dial out the hotline number automatically after the designated period of time.

To configure the hotline number via phone user interface:

- 1. Press Menu->Features->Others->Hot Line.
- 2. Enter the desired number in the Hotline Number field.
- 3. Enter the delay time in the Hotline Delay field.

. Number:		1234	
. Hotline Del	lay:	4	

The valid values range from 0 to 10 (seconds) and the default value is "4".

4. Press the **Save** soft key to accept the change or the **Back** soft key to cancel.

Hotline is configurable via web user interface at the path **Features**->**General Information**.

Basic Call Features

The SIP-T46G IP phone is designed to be easily used like a regular phone on a public switched telephone network (PSTN). You can place calls, answer calls, transfer a call to someone else, or conduct a conference call.

This chapter provides basic operating instructions for the SIP-T46G IP phone. Topics include:

- Placing Calls
- Answering Calls
- Ending Calls
- Redialing Numbers
- Recent Call In Dialing
- Auto Answer
- Auto Redial
- Call Completion
- ReCall
- Call Mute
- Call Hold/Resume
- Do Not Disturb (DND)
- Call Forward
- Call Transfer
- Call Waiting
- Conference
- Call Park
- Call Pickup
- Anonymous Call
- Anonymous Call Rejection

If you require additional information or assistance with your new phone, contact your system administrator.

Placing Calls

You can place a call in one of three ways using your SIP-T46G IP phone:

• Using the handset

- Using the speakerphone
- Using the headset

You can also dial the number first, and then choose the way you want to speak to the other party.

You can also search and dial a contact from call history, local directory or remote phone book. For more information, refer to Contact Management on page 61 and Call History Management on page 79.

During a call, you can alternate between Speakerphone, Headset, and Handset modes by pressing the Speakerphone key, the HEADSET key, or by picking up the handset.

The call duration of the active call and far-site's information (name or phone number) are visible on the LCD screen. In the figure below, the call to "Tom" (the phone number: 1021) has lasted 5 seconds.



To place a call using the handset:

- **1.** Pick up the handset.
- 2. Enter the desired number using the keypad.
- **3.** Press (o_{κ}) , $|| \#_{SENO} ||$, or the **Send** soft key.

The # key is configured as a send key by default. You can also set the * key as the send key, or set neither. For more information, refer to the Key As Send on page 52.

Note

You can also dial using the SIP URI or IP address. To obtain the IP address of a phone, press the **OK** key when the phone is idle. The maximum length of SIP URI or IP address length is 32 characters. For example, SIP URI: 3606@sip.com, IP: 192.168.1.15.

Your phone may not support direct IP dialing. Contact your system administrator for more information.

To place a call using the hands-free speakerphone mode:

Do one of the following:

- With the handset on-hook, press or the line key to obtain a dial tone.

Enter the desired number using the keypad.

Press $(\circ\kappa)$, $\#_{\text{send}}$ or the **Send** soft key.

-	With the	handset	on-hook,	enter the	desired	number	using	the ke	eypad.
---	----------	---------	----------	-----------	---------	--------	-------	--------	--------

Press , , #sevo , or the **Send** soft key.

To place a call using the headset:

Do one of the following:

With the optional headset connected, press is to activate the headset mode.
Press the line key to obtain a dial tone.
Enter the desired number using the keypad.
Press or , #**** , or the Send soft key.
With the optional headset connected, press or to activate the headset mode.

Enter the desired number using the keypad.

Press $(o\kappa)$, $\#_{send}$, or the **Send** soft key.

Note To permanently use the headset mode, refer to Headset Prior on page 83.

The SIP-T46G IP phone can handle multiple calls at a time. However, only one active call (the call that has audio associated with it) can be in progress at any time, other calls are placed on hold. The SIP-T46G IP phone can handle a maximum of 50 calls at one time.

To place multiple calls:

You can have more than one call on your SIP-T46G IP phone. To place a new call during an active call, do one of the following:

Press the line key. The active call is placed on hold.

Enter the desired number using the keypad.

- Press $(\mathbf{o}_{\mathsf{K}}), [[\texttt{\#}_{\mathtt{SEND}}]]$, or the **Send** soft key.
- Press (or the Hold soft key to place the original call on hold.

Press the New Call soft key.

Enter the desired number using the keypad.

Press (o_{κ}) , $\#_{SERO}$, or the **Send** soft key.

You can press (\bullet) or (\bullet) to switch between calls, and then press the **Resume** soft key to retrieve the desired call.

Note If multiple accounts are registered on the phone, you can first press the desired line key on the idle screen or press the **Line** soft key on the dialing screen, and then you can use the selected account to place a call.

Answering Calls

When you are not in another call, you can answer a call in one of three ways:

- Using the handset
- Using the speakerphone
- Using the headset
- **Note** You can reject incoming calls by pressing the **X** key or the **Reject** soft key. You can also activate Do Not Disturb mode to ignore the incoming calls without ringing on your phone. For more information, refer to Do Not Disturb (DND) on page 130.

You can forward incoming calls to someone else by pressing the **Forward** soft key. For more information, refer to Call Forward on page 134.

Answering When Not in Another Call

Call duration and destination will always appear on the LCD screen for the active call.

To answer a call using the handset:

1. Pick up the handset.

To answer a call using the hands-free speakerphone mode:

Do one of the following:

- Press .
- With the handset on-hook and the headset mode deactivated, press the Answer soft key.
- With the handset on-hook and the headset mode deactivated, press the line key with the fast flashing green LED indicator.

To answer a call using the headset:

Do one of the following:



- With the headset mode activated, press the Answer soft key.
- With the headset mode activated, press the line key with the fast flashing green LED indicator.

Answering When in Another Call

If you have an active call, and an incoming call arrives on the phone, do one of the following:

- Press the **Answer** soft key.

The incoming call is answered and the original call is placed on hold.

Press (•) to access the new call.

Press $(o\kappa)$ or the **Answer** soft key.

The incoming call is answered and the original call is placed on hold.

Ending Calls

To end a call:

Do one of the following:

- If you are using the handset, press (x) , the **End Call** soft key or hang up the handset.
- If you are using the headset, press (x), or the **End Call** soft key.
- If you are using the speakerphone, press (×) , 💷 or the **End Call** soft key.

Note

When a call is placed on hold, you cannot press (\times) to end it. You should press the **Resume** soft key to resume the call, and then press (\times) to end it.

Redialing Numbers

To redial the last dialed number from your phone:

1. Press 🕞 twice.

A call to your last dialed number is attempted.

To redial a previously dialed number from your phone:

- **1.** Press **(C)** when the phone is idle.
- Press or to select the desired entry from the placed calls list, and then press or the Send soft key.

Recent Call In Dialing

To view the placed calls list when the phone is on the dialing screen, you should enable recent call in dialing in advance.

To enable recent call in dialing via web user interface:

- 1. Click on Directory->Setting.
- 2. Select Enabled from the pull-down list of Recent Call In Dialing.

Yealink T466							Eng	Log Out Jish(English) 🗸
	Status	Account	Network	DSSKey	Features	Settings	Directory	Security
Local Directory Remote Phone Book Phone Call Info LDAP Multicast IP Setting	Direc	tory ? Disabled History Remote Pf LDAP ch Source List In Disabled Remote Pf LDAP	Dialing ?	Enabled Local Directory Enabled Local Directory History			the search sou the entered str results on the p Recent Call I It allows users calls list when the pre-dialing	d lists. e in Dialing P phone to search entries from ring, and display pre-dialing screen. n Dialing to view the placed the phone is on

3. Click **Confirm** to accept the change.

Note Recent call in dialing is configurable via web user interface only.

To view placed calls list when the phone is on the dialing screen:

1. Pick up the handset, press the Speakerphone key or press the line key.

The LCD screen displays the placed calls list.

% 1006			
<u>§</u> 1006			
	1001		Ī
	joy	5023	
	1056		
	sunmy	1002	
Directory	123		Cancel

⁶ 10		
sunmy	1002	
1056		l l
1001		

You can also enter a few continuous characters of the contact name or continuous numbers of the contact number (office, mobile or other number) to search from placed calls list.

Auto Answer

You can use auto answer feature to automatically answer an incoming call on a line. Auto answer is configurable on a per-line basis.

You can choose to mute the local microphone when an incoming call is answered automatically. For more information, refer to Keep Mute on page 128.

To configure auto answer via phone user interface:

- 1. Press Menu->Features->Auto Answer.
- 2. Select the desired line and then press the Enter soft key.
- **3.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select **Enabled** from the **Auto Answer** field.



4. Press the Save soft key to accept the change or the Back soft key to cancel.

The icon A appears on the LCD screen.

		A 14	1 53 Fri, Sep 02
1006			
			1
History	Directory	DND	Menu

Auto answer is configurable via web user interface at the path Account->Basic.

Note Auto answer is only applicable when there is no other call in progress on the phone.

Auto Redial

You can enable auto redial to automatically redial a phone number when the called party is busy. You can also configure the number of auto redial attempts and the time to wait between redial attempts.

To configure auto redial via phone user interface:

- 1. Press Menu->Features->Others->Auto Redial.
- 2. Press () or () , or the Switch soft key to select **Enabled** from the Auto Redial field.
- **3.** Enter the desired time (in seconds) in the **Redial Interval** field.

The default value is "10".

 Enter the desired number of redial attempts in the **Redial Times** field. The default value is "10".

	Auto	Redial	
1. Auto Redia	l:	Enabled	$\triangleleft \triangleright$
2. Redial Inte	rval:	10	
3. Redial Time	es:	10	
Back	123	Delete	Save

5. Press the Save soft key to accept the change or the Back soft key to cancel.

Auto redial is configurable via web user interface at the path Features->General Information.

To use auto redial:

When the called party is busy, the following prompt will appear on the LCD screen of the phone:



1. Press the **OK** soft key to activate auto redial. The following prompt will appear on the LCD screen of the phone:

ish	
1000	nes: 10, Interval: 8s
	Cancel

Wait for the designated period of time or press the OK soft key to redial the phone number.
 The phone will retry as many times as configured until the called party is idle.

Call Completion

You can use call completion to notify the caller who failed to reach a desired party when the party becomes available to receive a call.

To configure call completion via phone user interface:

- 1. Press Menu->Features->Others->Call Completion.
- 2. Press (•) or (•), or the Switch soft key to select Enabled from the Call Completion field.

Call Completion				
1. Call Completion:	Enabled	$\triangleleft \triangleright$		
Back	Switch	Save		

3. Press the Save soft key to accept the change or the Back soft key to cancel.

Call completion is configurable via web user interface at the path **Features**->**General Information**.

To use call completion:

When the called party is busy, the following prompt will appear on the LCD screen of the phone:

An .	13:34 Fri, Sep 02
Wait for 2021?	
1	ок
	A Wait for 2021?

1. Press the **OK** soft key, the phone returns to the idle screen and call completion is activated. When the called party becomes idle, the following prompt will appear on the LCD screen of the phone:

1006	A <u>n</u>	13:34 Fri, Sep 02
1006		
🕜 N	ote	
	Dialing 2021?	
Cancel		ОК

1. Press the **OK** soft key to redial the number.

Note Call completion is not available on all servers. For more information, contact your system administrator.

ReCall

You can press a recall key to place a call back to the last incoming call.

To configure a recall key via phone user interface:

- 1. Press Menu->Features->DSS Keys.
- 2. Select the desired DSS key.
- 3. Press (\cdot) or (\cdot) , or the Switch soft key to select **Key Event** from the **Type** field.
- 4. Press (•) or (•) , or the Switch soft key to select **ReCall** from the **Key Type** field.
- 5. (Optional.) Enter the string that will appear on the LCD screen in the Label field.

Line Key 6				
1. Type:	Key Event < 🖂			
2. Key Type:	ReCall	$\triangleleft \triangleright$		
3. Label:				
Back	Switch	Save		

6. Press the Save soft key to accept the change or the Back soft key to cancel.

ReCall key is configurable via web user interface at the path Dsskey->Line key.

Call Mute

You can mute the microphone of the active audio device during an active call so that the other party cannot hear you. You can also mute the microphone while dialing so that the other party cannot hear you when the call is set up. It helps prevent the other party from hearing the informal discussion when auto answer is enabled on his/her phone.

Normally, the mute state ends when the active call ends. Keep mute feature enables you to make the mute state of the phone persist across calls and to mute the phone's microphone on any screen (e.g., idle screen). It helps prevent the other party from hearing the noise coming from your room when auto answer is enabled on your phone.

Call mute applies to all modes (Handset, Headset and Speakerphone).

Mute a Call

To mute a call:

1. Press (*i*) during an active call.

The mute key LED illuminates solid red. The LCD screen indicates that the call is now muted.

() Talking			
<u>&</u> 1006	1000	Mute 01 : 12	
Transfer	Hold	Conference	End Call

To un-mute a call:

 Press again to un-mute the call. The mute key LED goes out.

Mute While Dialing

To mute the microphone while dialing:

1. Press () on the pre-dialing, dialing or calling screen.

The mute key LED illuminates solid red. The call is muted automatically when setting up successfully.

To un-mute the microphone while dialing:

1. Press 🕼 again on the pre-dialing, dialing or calling screen.

Note You can also mute the microphone when the IP phone is ringing.

Keep Mute

If keep mute is enabled, you can mute the microphone on any screen. The phone stays in the mute state until you un-mute the microphone or until the phone restarts. This feature should be pre-configured by your system administrator.

To mute the microphone on any screen:

1. Press Ø on any screen (for example, ringing, dialing, talking or idle).

1006		ال 🖇 🖇	8:39 Fri, Sep 02
<u>1006</u>			
History	Directory	DND	Menu

The mute key LED illuminates solid red. The icon $\oint \int$ appears on the status bar.

To un-mute the microphone on any screen:

Press again on any screen (for example, ringing, dialing, talking or idle).
 The mute key LED goes out. The icon ⁶/₂ disappears from the status bar.

Call Hold/Resume

You can place an active call on hold. Only one active call can be in progress at any time. Other calls can be made and received while placing the original call on hold. When you place a call on hold, your IP PBX may play music to the other party while waiting.

To place a call on hold:

1. Press or the Hold soft key during a call.

The line key LED flashes green. The LCD screen indicates that the call is on hold.

◀) Hold			HD 🛿
<u>&</u> 1025	1026	U Hold 00 : 04	
Transfer	Resume	New Call	End Call

Note The phone will beep softly every 30 seconds to remind you that you still have a call on hold. The phone ignores engaged audio device (handset or headset) and switches to Speakerphone to play the hold tone by default. The Speakerphone key will light up for 1 second.

To resume a held call:

1. Press or the **Resume** soft key.

Multiple Calls on Hold:

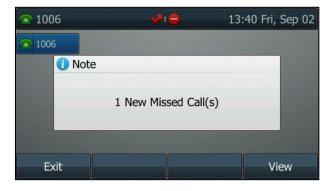
If multiple calls are placed on hold, do one of the following:

- Press or to switch between the calls, and then press the **Resume** soft key to retrieve the desired call.
- Press the corresponding line key to retrieve the call.

If more than two calls are on hold, a numbered prompt appears on the LCD screen, for example "2/4", indicating that this is the second call out of four calls.

Do Not Disturb (DND)

You can use DND to reject incoming calls automatically on the phone. The prompt message "**n New Missed Call(s)**" ("n" indicates the number of missed calls, e.g., 1 New Missed Call(s)) will appear on the LCD screen, and callers will receive a busy message. All calls you receive while DND is enabled are logged to your missed calls list.





When DND feature is enabled, the IP phone supports displaying a large DND icon on the idle screen. For more information, contact your system administrator.

Note The prompt message will display only if Missed Call Log for the line is enabled. Missed call log is configurable via web user interface at the path **Account**->**Basic**.

Do not disturb is local to the phone, and may be overridden by the server settings. For more information, contact your system administrator.

You can enable/disable DND for the phone system, or you can customize DND for each or all accounts. Two DND modes:

- **Phone** (default): DND is effective for the phone system.
- **Custom**: DND can be configured for each or all accounts.

You can receive incoming calls from authorized numbers when DND is enabled.

To configure the DND mode via web user interface:

- 1. Click on Features->Forward&DND.
- 2. In the DND block, mark the desired radio box in the Mode field.

Yealink 1466							Engl	Log Out ish(English) 🗸
	Status	Account	Network	DSSKey	Features	Settings	Directory	Security
Forward&DND	F	orward					NOTE	
General Information		Forward Emergen Forward Authorize		Disabled	((Call Forward It allows users	
Audio		Mode		● Phone ○ Cu			incoming call to Call Forward	Mode
Intercom		Account		1006	× (Phone: Call for effective for the Custom: Call for	e IP phone. orward feature
Transfer				:			accounts.	ed for each or all
Call Pickup				•			Do Not Distur It allows IP pho incoming calls.	b (DND) ones to ignore
Remote Control Phone Lock								ature is effective
ACD	U	DND Construction		Disabled	✓ (for the IP phon Custom: DND configured for e	feature can be
SMS		DND Authorized N	umbers				accounts.	rk here to get
Action URL		Mode		Phone O Cu 1006	istom 🕜		more guides.	ck here to get
Bluetooth		DND Status		○ On ⑧ Off	0			
Power LED		DND On Code						
Notification Popups		DND Off Code	rm		Cancel			

- 3. Click **Confirm** to accept the change.
- Note DND mode is configurable via web user interface only.

To enable DND in phone mode:

- 1. Press the DND soft key when the phone is idle.
 - The icon 😑 appears on the status bar.



Note When DND and busy forward are enabled in phone mode, calls will be sent to the configured destination number. For more information on busy forward, refer to Call Forward on page 135.

To enable DND in custom mode:

1. Press the DND soft key when the phone is idle.

The LCD screen displays a list of accounts on the phone.

	Custor	m DND	
1. 1006:		Off	< ►
2. 1007:		Off	$\triangleleft \triangleright$
Back	All On	Enter	Save
Dack	All Off	Enter	Save

- **2.** Press (\bullet) or (\bullet) to select the desired account, and then press the **Enter** soft key.
- **3.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select **Enabled** from the **DND Status** field.
- (Optional.) Enter the DND on code or off code respectively in the On Code or Off Code field.

If the on code or off code is configured, the IP phone will send the corresponding code to activate or deactivate DND on the server. It is not available on all servers.

5. Press the Save soft key to accept the change.

If you want to enable DND for all accounts, you can press the All On soft key.

If you enable DND for the default account, the associated line icon will change to 2, and the icon 2 will appear on the status bar.

1006		9 1	3:39 Fri, Sep 02
1006			
1007			
History	Directory	DND	Menu

If you activate DND for the non-default account, only the associated line icon will change to \mathbf{T} .

1007		13	:39 Fri, Sep 02
1006			
1007			
History	Directory	DND	Menu

If you enable DND for all accounts, all line icons will change to \bigcirc and the icon \bigcirc will appear on the status bar.



Note When DND and busy forward are both enabled for a specific account, calls to the specific account will be sent to the configured destination number. For more information on call forward, refer to Call Forward on page 134.

To configure the DND authorized numbers via web user interface:

1. Click on Features->Forward&DND.

- 2. Select Enabled from the pull-down list of DND Emergency.
- 3. Enter the numbers in the DND Authorized Numbers field.

For multiple numbers, enter a comma between every two numbers.

Yealink 1466			Log Out English(English) V
	Status Account Network	DSSKey Features	Settings Directory Security
Forward&DND General Information Audio Intercom Transfer Call Pickup	Forward Forward Emergency Forward Authorized Numbers Mode Account	Disabiled ✓ ? ? ? 1006 ✓ ? . . .	NOTE Call Forward It allows users to redirect an incoming call to a third party. Call Forward Mode Phone: Call forward feature is effective for the IP phone. Custom: Call forward feature can be configured for each or all accounts. Do Not Disturb (DND) It allows IP phones to inorce
Remote Control Phone Lock	DND 🕜		incoming calls. DND Mode Phone: DND feature is effective for the IP phone.
ACD	DND Emergency DND Authorized Numbers	Enabled V 🖓	Custom: DND feature can be configured for each or all accounts.
Action URL	Mode Account	 ○ Phone ● Custom ② 1006 ✓ ② 	You can click here to get more guides.
Bluetooth	DND Status	○ On ● Off 🕜	
Power LED Notification Popups	On Code Off Code		
	Confirm	Cancel	

4. Click **Confirm** to accept the change.

When DND is enabled on the phone, the phone can still receive incoming calls from the numbers specified in the **DND Authorized Numbers** field.

Note

DND authorized number is configurable via web user interface only.

When the phone misses a call, a prompt window will pop up by default. If you want to disable the feature, contact your system administrator for more information.

To disable DND in phone mode:

1. Press the **DND** soft key when the phone is idle.

To disable DND in custom mode:

- 1. Press the **DND** soft key when the phone is idle.
- 2. Press () or () to select the desired account and then press the **Enter** soft key.
- 3. Press (\cdot) or (\cdot) , or the Switch soft key to select Disabled from the DND Status field.
- 4. Press the **Save** soft key to accept the change.

If you want to disable DND for all accounts, press the All Off soft key.

Call Forward

You can configure your phone to forward incoming calls to another party (static forwarding). You can also forward calls while your phone is ringing (dynamic forwarding).

Note

When the phone forwards a call, a prompt window will pop up by default. If you want to disable the feature, contact your system administrator for more information.

Static Forwarding

Three types of static forwarding:

- Always Forward: Incoming calls are immediately forwarded.
- Busy Forward: Incoming calls are immediately forwarded if the phone is busy.
- No Answer Forward: Incoming calls are forwarded if not answered after a period of time.

You can enable/disable call forward for the phone system, or you can customize call forward for each or all accounts. Two call forward modes:

- **Phone** (default): Call forward is effective for the phone system.
- **Custom:** Call forward can be configured for each or all accounts.

To configure the call forward mode via web user interface:

- 1. Click on Features->Forward&DND.
- 2. In the Forward block, mark the desired radio box in the Mode field.

Yealink 1466			Log Out English(English) V
	Status Account Network	DSSKey Features	Settings Directory Security
Forward&DND	Forward		NOTE
General Information	Forward Emergency Forward Authorized Numbers	Disabled V	Call Forward It allows users to redirect an incoming call to a third party.
Audio	Mode	Phone O Custom ??	Call Forward Mode
Intercom	Account Always Forward	1006 ∨ ?	Phone: Call forward feature is effective for the IP phone. Custom: Call forward feature
Transfer	Target		can be configured for each or all accounts.
Call Pickup	On Code		Do Not Disturb (DND) It allows IP phones to ignore
Remote Control	Off Code	0	incoming calls.
Phone Lock	Busy Forward	🔾 On 🖲 Off 🕜	DND Mode Phone: DND feature is effective for the IP phone.
ACD	Target On Code		Custom: DND feature can be configured for each or all accounts.
SMS	Off Code	0	You can click here to get
Action URL	No Answer Forward	🔿 On 🖲 Off 🕜	more guides.
Bluetooth	After Ring Time(0~120s)	12 🗸 🖉	
Power LED	Target	0	
101101 220	On Code	0	
Notification Popups	Off Code	0	

3. Click **Confirm** to accept the change.

Note Call forward mode is configurable via web user interface only.

To enable call forward in phone mode:

- 1. Press Menu->Features->Call Forward.
- **2.** Press (\bullet) or (\bullet) to select the desired forwarding type, and then press the **Enter** soft key.
- 3. Depending on your selection:
 - a) If you select Always Forward:
 - Press (•) or (•) , or the Switch soft key to select Enabled from the Always Forward field.
 - Enter the destination number you want to forward all incoming calls to in the Forward to field.
 - (Optional.) Enter the always forward on code or off code respectively in the On Code or Off Code field.

	Alway	s Forward	
1. Always Forward:		Enabled	$\triangleleft \triangleright$
2. Forward to:		1234	
3. On Code:		*71	
4. Off Code:		*72	
Back	123	Delete	Save

- b) If you select Busy Forward:
 - Press (•) or (•), or the Switch soft key to select Enabled from the Busy Forward field.
 - **2)** Enter the destination number you want to forward incoming calls to when the phone is busy in the **Forward to** field.
 - (Optional.) Enter the busy forward on code or off code respectively in the On Code or Off Code field.

	Busy I	Forward	
1. Busy Forward:		Enabled < D	
2. Forward to:		1234	
3. On Code:		*73	
4. Off Code:		*74	
Back	123	Delete	Save

- c) If you select No Answer Forward:
 - Press (•) or (•), or the Switch soft key to select Enabled from the No Answer Forward field.
 - Enter the destination number you want to forward unanswered incoming calls to in the Forward to field.
 - Press or , or the Switch soft key to select the ring time to wait before forwarding from the After Ring Time field.

The default ring time is 12 seconds.

4) (Optional.) Enter the no answer forward on code or off code respectively in the On Code or Off Code field.

	No Ans	wer Forward	
1. No Answer Forward:		Enabled	$\triangleleft \triangleright$
2. Forward to:		1234	
3. After Ring Time:		12s	∢ ►
4. On Code:	4. On Code:		
5. Off Code:		*76	
Back	123	Delete	Save

4. Press the Save soft key to accept the change or the Back soft key to cancel.

The icon \bigcirc on the status bar indicates that the call forward is enabled.

To enable call forward in custom mode:

- 1. Press Menu->Features->Call Forward.
- **2.** Press (\bullet) or (\bullet) to select the desired account, and then press the **Enter** soft key.

	Custom Forward	
1. 1006:	Off	
2. 1007:	Off	
Back		Enter

- **3.** Press \frown or \bigcirc to select the desired forwarding type, and then press the **Enter** soft key.
- 4. Depending on your selection:
 - a) If you select Always Forward, you can enable it for a specific account.
 - Press or , or the Switch soft key to select Enabled from the Always Forward field.
 - 2) Enter the destination number you want to forward all incoming calls to in the

Forward to field.

 (Optional.) Enter the always forward on code or off code respectively in the On Code or Off Code field.

	Alway	s Forward	
1. Always Forv	vard:	Enabled	$\triangleleft \triangleright$
2. Forward to:		1234	
3. On Code:		*71	
4. Off Code:		*72	
4. Off Code.		*72	
Back	123	Delete	Save

You can also enable always forward for all accounts. After always forward was enabled for a specific account, do the following:

- **1)** Press (\bullet) or (\bullet) to highlight the **Always Forward** field.
- 2) Press the All Lines soft key.

The LCD screen prompts "Copy to all lines?".

	Always Forward	
1. Always Forward:	Enabled	
2. Foi 🚺 Note		0.05523/017
3. On 4. Off	Copy to all lines?	
Cancel		ОК

- 3) Press the **OK** soft key to accept the change or the **Cancel** soft key to cancel.
- b) If you select Busy Forward, you can enable it for a specific account.
 - Press or , or the Switch soft key to select Enabled from the Busy Forward field.
 - **2)** Enter the destination number you want to forward incoming calls to when the phone is busy in the **Forward to** field.

 (Optional.) Enter the busy forward on code or off code respectively in the On Code or Off Code field.

	Busy	Forward	
1. Busy Forwar	rd:	Enabled	∢ ►
2. Forward to:		1234	
3. On Code:		*73	
4. Off Code:		*74	
Back	123	Delete	Save

You can also enable busy forward for all accounts. After busy forward was enabled for a specific account, do the following:

- **1)** Press (\bullet) or (\bullet) to highlight the **Busy Forward** field.
- 2) Press the All Lines soft key.

The LCD screen prompts "Copy to all lines?".

	Busy Forward	
1. Busy Forward:	Enabled	< >
2. Fol () Note 3. On 4. Off	Copy to all lines?	
Cancel		ОК

- 3) Press the **OK** soft key to accept the change or the **Cancel** soft key to cancel.
- c) If you select No Answer Forward, you can enable it for a specific account.
 - 1) Press or , or the Switch soft key to select Enabled from the No Answer Forward field.
 - Enter the destination number you want to forward unanswered incoming calls to in the Forward to field.
 - Press or , or the Switch soft key to select the ring time to wait before forwarding from the After Ring Time field.

The default ring time is 12 seconds.

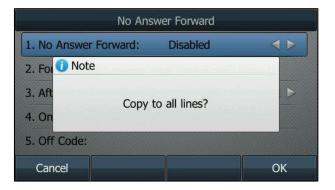
4) (Optional.) Enter the no answer forward on code or off code respectively in theOn Code or Off Code field.

	No Ans	wer Forward	
1. No Answer	Forward:	Enabled	$\triangleleft \triangleright$
2. Forward to	:	1234	
3. After Ring	Time:	12s	<►
4. On Code:		*75	
5. Off Code:		*76	
Back	123	Delete	Save

You can also enable no answer forward for all accounts. After no answer forward was enabled for a specific account, do the following:

- **1)** Press (\bullet) or (\bullet) to highlight the **No Answer Forward** field.
- 2) Press the All Lines soft key.

The LCD screen prompts "Copy to all lines?".



3) Press the **OK** soft key to accept the change or the **Cancel** soft key to cancel.

5. Press the Save soft key to accept the change or the Back soft key to cancel.

If you enable call forward for the default account, the associated line icon will change to 2, and the icon 2 will appear on the status bar.



If you enable call forward for the non-default account, only the associated line icon will change to $rac{rac}{2}$.

<u>र</u> 1006		13	:50 Fri, Sep 02
1006			
6 1007			
History	Directory	DND	Menu

If you enable call forward for all accounts, all line icons will change to 5, and the icon 5 will appear on the status bar.

1007		5	13:49 Fri,	Sep 02
1006				
1007				
History	Directory	DND	M	enu

If the forward on code or off code is configured, the IP phone sends the corresponding code to the server when a certain action takes place. For example, if you set the Always Forward on code to be *71 (may vary on different servers), and the target number to be 1234. When you enable Always Forward on the IP phone, the IP phone sends *711234 to the server, and then the server will enable Always Forward feature on the server side, hence being able to get the right status of the extension.

Call forward is configurable via web user interface at the path Features->Forward&DND.

Note

You can also enter the SIP URI or IP address in the **Forward to** field. For more information on using the SIP URI or IP address, refer to Placing Calls on page 117.

Call forward is local to the phone, and may be overridden by the server settings. Call forward on code or off code may be different between servers. For more information, contact your system administrator.

To configure the forward authorized numbers via web user interface:

- 1. Click on Features->Forward&DND.
- 2. Select Enabled from the pull-down list of Forward Emergency.
- 3. Enter the numbers in the Forward Authorized Numbers field.

ealink 146G							English(English)
	Status	Account	Network	DSSKey	Features	Settings	Directory Security
Forward&DND	1	Forward					NOTE
General		Forward Emergence		Enabled	✓ 0		Call Forward
Information		Forward Authorize	d Numbers	1002,1008	0		It allows users to redirect an incoming call to a third party.
Audio		Mode		🔿 Phone 🖲 Cu	stom 🕜		Call Forward Mode
Intercom		Account		1006	✓ ?		Phone: Call forward feature is effective for the IP phone.
Intercom		Always Forward		◯ On ◉ Off	0		Custom: Call forward feature can be configured for each or a
Transfer		Target			0		accounts.
Call Pickup		On Code			0		Do Not Disturb (DND) It allows IP phones to ignore
Remote Control		Off Code			0		incoming calls.
Phone Lock		Busy Forward		○ On ⑧ Off	0		DND Mode Phone: DND feature is effectiv
		Target			0		for the IP phone. Custom: DND feature can be
ACD		On Code			0		configured for each or all accounts.
SMS		Off Code			0		You can click here to get
Action URL		No Answer Forw	ard	○ On ⑧ Off	0		more guides.
Bluetooth		After Ring Time	e(0~120s)	12	✓ 🕜		
		Target			0		
Power LED		On Code			0		
Notification		Off Code			0		

For multiple numbers, enter a comma between every two numbers.

4. Click **Confirm** to accept the change.

When call forward is enabled on the phone, the phone cannot forward incoming calls from the numbers specified in the **Forward Authorized Numbers** field.

Note Forward authorized number is configurable via web user interface only.

To disable call forward in phone mode:

Do one of the following:

- Press **[**[+] when the phone is idle.
- Press Menu->Features->Call Forward.

Press (\bullet) or (\bullet) to select the desired forwarding type and then press the **Enter** soft key.

Press (\cdot) or (\cdot) , or the **Switch** soft key to select **Disabled** to disable the call forward.

Press the Save soft key to accept the change.

To disable call forward in custom mode:

- 1. Press Menu->Features->Call Forward or press ((+()) when the phone is idle.
- **2.** Press (\bullet) or (\bullet) to select the desired account and then press the **Enter** soft key.
- 3. Press () or () to select the desired forwarding type and then press the **Enter** soft key.
- **4.** Press (•) or (•), or the **Switch** soft key to select **Disabled** to disable the call forward.

If you want to disable call forward for all accounts, press the **All Lines** soft key.

5. Press the Save soft key to accept the change.

Dynamic Forwarding

You can manually forward a call to another number while your phone rings.

To forward an incoming call to another party:

- 1. When the phone is ringing, press the Forward soft key.
- 2. Enter the number you want to forward the incoming call to.

(I) Forwar	d to			
<u>§</u> 1006	123			
	dd		1233	
	dd		122323	μ
	dd		44546	
Send		123	Delete	Cancel

3. Press (or , #sero) , or the Send soft key.

The LCD screen prompts a call forward message.

Call Transfer

You can transfer a call to another party during a call in one of three ways:

- Blind Transfer: Transfer a call directly to another party without consulting.
- Semi-Attended Transfer: Transfer a call when the target phone is ringing.
- Attended Transfer (Consultative Transfer): Transfer a call with prior consulting.

Blind Transfer

Before performing a blind transfer (except you use dial number method), make sure that the transfer mode on DSS key is configured as **Blind Transfer**. For more information on how to configure it, refer to Configuring Transfer Mode on DSS Key on page 167.

To perform a blind transfer during a call:

- 1. Press ((+()) or the Transfer soft key during a call.
- 2. You can do one of the following:

- Enter the number you want to transfer the call to.

(I) Transfe	r to		2/2
<u>§</u> 1006	123		
	dd	1233	
	dd	122323	H
	dd	44546	
Transfer	123	Delete	Cancel

Press or the **Transfer** soft key to complete the transfer.

If you are using a handset, the transfer can be completed by hanging up the handset.

- Press the **Directory** soft key, and then select **Local Directory**. Select the desired group and search for the contact (Directory should be configured in advance. Refer to Directory on page 61 for more information).
 - Press $(\mathbf{o}\mathbf{k})$, $||\mathbf{#}_{sevo}||$ or the **Send** soft key to complete the transfer.
- Press the **Directory** soft key, and then select **History**. Select the desired list and then press or or to select the entry (Directory should be configured in advance. Refer to Directory on page 61 for more information).
 - Press $(\mathbf{o}\mathbf{K})$, $[\mathbf{\#}_{seno}]$ or the **Send** soft key to complete the transfer.
- Press the **Directory** soft key, and then select **Remote Phone Book**. Select the desired group and search for the contact (Directory should be configured in advance. Refer to Directory on page 61 and Remote Phone Book on page 77 for more information).
 - Press (o_{K}) , $\#_{sevo}$ or the **Send** soft key to complete the transfer.

To perform a blind transfer to exist call:

- 1. Press ((+) or the **Transfer** soft key during a call.
- **2.** Press $\overbrace{\frown}$ or (\frown) to select the desired hold call you want to transfer the call to.

📞 Transfer	• to		4/4
§ 1006			
	Hold 1001	1001	
	Hold 1048	1048	
		26	
Transfer	123	Retrieve	Cancel

3. Press (o_{κ}) , $\#_{sevo}$ or the **Transfer** soft key to complete the transfer.

Then the call is connected to the number to which you are transferring.

Semi-attended Transfer

Before performing a semi-attended transfer (except you use dial number method), make sure that the transfer mode on DSS key is configured as **New Call** or **Attended Transfer**. For more information on how to configure it, refer to Configuring Transfer Mode on DSS Key on page 167.

To perform a semi-attended transfer:

- **1.** Press ((-()) or the **Transfer** soft key during a call.
- 2. Do one of the following:
 - Enter the number you want to transfer the call to.

Press ок or #seno to dial out.

- Press the **Directory** soft key, and then select **Local Directory**. Select the desired group and search for the contact (Directory should be configured in advance. Refer to Directory on page 61 for more information).
 - Press (or, (#seno) or the **Send** soft key to dial out.
- Press the **Directory** soft key, and then select **History**. Select the desired list and then press or or to select the entry (Directory should be configured in advance. Refer to Directory on page 61 for more information).
 - Press $(\mathbf{o}_{\mathbf{K}})$, $[\mathbf{\#}_{\mathtt{sevo}}]$ or the **Send** soft key to dial out.
- Press the **Directory** soft key, and then select **Remote Phone Book**. Select the desired group and search for the contact (Directory should be configured in advance. Refer to Directory on page 61 and Remote Phone Book on page 77 for more information).
 - Press $(\mathbf{v}, \mathbf{H}_{sse})$ or the **Send** soft key to dial out.
- **3.** Press **(+(**) or the **Transfer** soft key to complete the transfer when receiving ringback. If you are using a handset, the transfer can be completed by hanging up the handset.

Attended Transfer

Before performing an attended transfer (except you use dial number method), make sure that the transfer mode on DSS key is configured as **New Call** or **Attended Transfer**. For more information on how to configure it, refer to Configuring Transfer Mode on DSS Key on page 167.

To perform an attended transfer:

- **1.** Press ((-()) or the **Transfer** soft key during a call.
- 2. Do one of the following:
 - Enter the number you want to transfer the call to.

Press (or) or (#seno) to dial out.

 Press the **Directory** soft key, and then select **Local Directory**. Select the desired group and search for the contact (Directory should be configured in advance. Refer to Directory on page 61 for more information).

```
Press (\mathbf{o}_{\mathbf{K}}), (\mathbf{\#}_{\mathtt{sevo}}) or the Send soft key to dial out.
```

- Press the **Directory** soft key, and then select **History**. Select the desired list and then press (•) or (•) to select the entry (Directory should be configured in advance. Refer to Directory on page 61 for more information).
 - Press (o_{κ}) , $[\#_{sevo}]$ or the **Send** soft key to dial out.
- Press the **Directory** soft key, and then select **Remote Phone Book**. Select the desired group and search for the contact (Directory should be configured in advance. Refer to Directory on page 61 and Remote Phone Book on page 77 for more information).
 - Press $(\circ\kappa)$, $[\#_{seno}]$ or the **Send** soft key to dial out.
- **3.** After the party answers the call, press **(-(**) or the **Transfer** soft key to complete the transfer.

If you are using a handset, the transfer can be completed by hanging up the handset. You can cancel the transfer before the call is connected by pressing the **Cancel** soft key.

Call Waiting

You can enable or disable call waiting on the phone. If call waiting is enabled, you can receive another call while there is already an active call on the phone. Otherwise, another incoming call is automatically rejected by the phone with a busy message when there is an active call on the phone. You can also enable or disable the phone to play a warning tone when receiving another call.

To configure call waiting via phone user interface:

- 1. Press Menu->Features->Call Waiting.
- 2. Press () or (), or the Switch soft key to select Enabled from the Call Waiting field.
- 3. Press (\cdot) or (\cdot) , or the Switch soft key to select Enabled from the Play Tone field.
- (Optional.) Enter the call waiting on code or off code respectively in the On Code or Off Code field.

1. Call Waiting:	Enabled	< >
2. Play Tone:	Enabled	< >
3. On Code:	 1	
4. Off Code:		

If on code or off code is configured, the IP phone will send the corresponding code to activate or deactivate call waiting on the server. It is not available on all servers.

5. Press the Save soft key to accept the change or the Back soft key to cancel.

Call waiting is configurable via web user interface at the path Features->General Information.

Note If on code or off code is not configured, the call waiting may not take effect. It depends on your server. Contact your serve administrator r for more information.

Conference

You can create a conference with other two parties using the phone's local conference. You can create a conference between an active call and a call on hold by pressing the Conference soft key. The SIP-T46G also supports network conference.

Note

Network conference is not available on all servers. For more information, contact your system administrator.

Local Conference

The SIP-T46G IP phone supports up to 3 parties (including yourself) in a conference call. This is the default method of conference called Local Conference.

Note To set up a local conference call, make sure local conference feature is enabled. For more information, contact your system administrator.

To set up a local conference call:

- **1.** Place a call to the first party.
- When the first party answers the call, press the Conference soft key to place a new call. The active call is placed on hold.
- **3.** Enter the number of the second party and then press (o_k) , $|| \#_{evo} ||$, or the **Send** soft key.

4. When the second party answers the call, press the **Conference** soft key again to join all parties in the conference.

📞 Confere	ence		
<u>&</u> 1006	3606 1000	00 : 12	
Split	Hold	Manage	End Call

To join two calls in a conference:

- **1.** Place two calls using two different accounts on the phone (for example, place the first call using account 1, and then place the second call using account 2).
- 2. Press (•) or (•) to select the call for conference and ensure that the call is active (for example, select the call on account 1).
- **3.** Press the **Conference** soft key to join the two calls in the conference on the selected account.

During the conference call, you can do the following actions:

- Press 👧 or the **Hold** soft key to place the conference on hold.
- Press the Split soft key to split the conference call into two individual calls on hold.
- Press the **Manage** soft key, and then press (\bullet) or (\bullet) to select the desired party:
 - Press the **Far Mute** soft key to mute the party. The muted party can hear everyone, but no one can hear the muted party.
 - Press the **Remove** soft key to remove the party from the conference call.
 - Press the **New Call** soft key to place a new call.
 - Press the **Back** soft key to return to the previous screen.
- Press for mute the conference call, all other participants can hear each other, but they cannot hear you.
- Press the **End Call** soft key to drop the conference call.

Network Conference

You can use network conference feature on the SIP-T46G IP phone to conduct a conference with multiple participants.

This feature allows you to perform the following:

• Join two calls together into a conference call.

• Invite another party into an active conference call.

To use this feature, contact your system administrator for the network conference URI in advance.

To configure network conference via web user interface:

- 1. Click on Account->Advanced.
- 2. Select the desired account from the pull-down list of Account.
- 3. Select Network Conference from the pull-down list of Conference Type.
- 4. Enter the conference URI (e.g., conference@example.com) in the Conference URI field.

Yealink 1466			Log Out English(English) –
	Status Account Network	DSSKey Features Settings	Directory Security
Register	Account	Account 1 🔹 ?	NOTE
Basic	Keep Alive Type	Default 🗸 🕜	DTME
Basic	Keep Alive Interval(Seconds)	30	DTMF It is the signal sent from the IP
Codec	RPort	Disabled 🗸 🕜	phone to the network, which is generated when pressing the IP
Advanced	Subscribe Period(Seconds)	1800	phone's keypad during a call.
	DTMF Type	RFC2833 - 🕜	Session Timer
	DTMF Info Type	DTMF-Relay	It allows a periodic refresh of SIP sessions through a
		•	re-INVITE request, to determine whether a SIP
		:	session is still active.
			Busy Lamp Field/BLF List
	SIP Registration Retry Timer(0~1800s)	30	Monitors a specific extension/a list of extensions for status
	Conference Type	Network Conference 👻 🕜	changes on IP phones.
	Conference URI	conference@example.com	Shared Call Appearance
	ACD Subscribe Period(120~3600s)	3600	(SCA)/ Bridge Line Appearance (BLA)
	Out Dialog BLF	Disabled 🗸 🍞	It allows users to share a SIP line on several IP phones. Any
	VQ RTCP-XR Collector name	0	IP phone can be used to originate or receive calls on the
	VQ RTCP-XR Collector address	0	shared line.
	VQ RTCP-XR Collector port	5060	Network Conference
	Confirm	Cancel	It allows multiple participants (more than three) to join in a call

5. Click **Confirm** to accept the change.

Note

Network conference is configurable via web user interface only.

To set up a network conference call:

- **1.** Place a call to the first party.
- Press the Conference soft key to place a new call. The active call is placed on hold.
- **3.** Enter the number of the second party and then press $(\circ\kappa)$, $[\#_{seco}]$, or the **Send** soft key.
- **4.** When the second party answers the call, press the **Conference** soft key to add the second party to the conference.
- 5. Press the Conference soft key to place a new call.

The conference is placed on hold.

- **6.** Enter the number of the new party and then press (∞) , $[\#_{\text{sevo}}]$, or the **Send** soft key.
- 7. When the new party answers the call, press the Conference soft key to add the new party

to the conference.

8. Repeat steps 5 to 7 until you have added all intended parties.

The procedures to set up a network conference call on specific servers may be different from introduced above. Contact your system administrator for more information.

Call Park

You can use call park feature to place a call on hold, and then retrieve the call from another phone in the system (for example, a phone in another office or conference room). You can park an active call by pressing the **Park** soft key or call park key (if configured) on the phone. If the call is parked successfully, there is a voice prompt confirming that the call was parked. You can retrieve the parked call by pressing the **Retrieve** soft key or the retrieve key (if configured). If the parked call is not retrieved within a period of time defined by the system, the phone performing call park will receive the call back.

Note Call park is not available on all servers. Contact your system administrator for more information.

The IP phone supports call park feature under the following modes:

- **FAC mode**: park the call to the local extension or a desired extension through dialing the park code, and retrieve the parked call from the local extension or desired extension through dialing the park retrieve code.
- **Transfer mode**: park the call to shared parking lot through performing a blind transfer to a call park number (call park code), and retrieve the parked call from the shared parking lot through dialing the park retrieve code.

You may need to configure the call park code and park retrieve code before using call park feature. You can also configure a call park key and retrieve key. For more information, refer to DSS Keys on page 90.

Note The call park code and park retrieve code are predefined on the system server. Contact your system administrator for more information.

FAC Mode

To configure call park feature in FAC mode via web user interface:

- 1. Click on Features->Call Pickup.
- 2. Select FAC from the pull-down list of Call Park Mode.
- 3. Select Enabled from the pull-down list of Call Park.

If **Enabled** is selected, the **Park** soft key will display on the LCD screen during a call, and the **Retrieve** soft key will display on the dialing screen.

4. (Optional.) Enter the call park code in the Call Park Code field.

It is configured for the **Park** soft key. This call park code will also apply to the call park key.

5. (Optional.) Enter the park retrieve code in the **Park Retrieve Code** field.

It is configured for the **Retrieve** soft key. This park retrieve code will also apply to the retrieve key.

									Log Out
Yealink 146G								Eng	ish(English) 🗸
	Status	Account	Network	DSSKey	Featur	es	Settings	Directory	Security
Forward&DND	C	all Pickup						NOTE	
General		Directed Call Picku Directed Call Picku		Enabled	•	0		Directed Call Picks up an inc	
Information Audio		Group Call Pickup	ip code	Enabled	•	0		specific extens	ion.
Intercom		Group Call Pickup	Code	*98		0		Directed Call Picks up incom pre-defined gro	ing calls within a
Transfer		Visual Alert for BL Audio Alert for BL		Enabled	•	0		You can config directed/group	
Call Pickup		Ring Type for BLF		Splash.wav	•	Ŭ		feature for the Visual Alert for	IP phone.
Remote Control	C	all Park 🕜						It allows the su to display a visi	pervisor's phone al prompt when
Phone Lock		Call Park Mode Call Park		FAC	•	0		incoming call.	user receives an
ACD		Call Park Code		*68		0			or BLF Pickup pervisor's phone tone when the
SMS		Park Retrieve Cod	e	*88		0		monitored use incoming call.	
Action URL		Confi	rm		Cancel			You can cli more guides.	ck here to get
Divotooth								more guides.	

6. Click **Confirm** to accept the change.

Note If the Park or Retrieve soft key doesn't appear on the LCD screen, please select Disabled from the pull-down list of Custom Softkey via web user interface at path Settings->Softkey Layout.

To park a call in FAC mode:

- During a call, press the **Park** soft key (You may need to press the **More** soft key to see the **Park** soft key) or the call park key (if configured).
 - If the call park code is not configured, you need to enter the call park code.

4603	*68	
		-
		000000000000000000000000000000000000000

Press $(\circ\kappa)$, $\#_{sevo}$ or the **Park** soft key.

- If the call park code is configured, the phone will dial the configured call park code shown as below:



- 2. Do one of the following:
 - a) If you want to park the call against the local extension.
 - **1)** Press **(#**_{SEND}**)**.

If the call is parked successfully, you will hear a voice prompt confirming that the call is parked.

- **b)** If you want to park the call against desired extension.
 - 1) Enter an extension (e.g., 4605) where you want to park the call.
 - 2) Press (ок) or (#_{send}).

If the call is parked successfully, you will hear a voice prompt confirming that the call is parked. The call is parked against the extension you entered.

To retrieve a parked call in FAC mode:

- **1.** Do one of the following:
 - If the park retrieve code is not configured, dial the park retrieve code (e.g., *88).
 - If the park retrieve code is configured, press the **Retrieve** soft key on the dialing screen or retrieve key (if configured) on the IP phone.

The phone will dial the configured park retrieve code and the Retrieve screen appears as below:



- 2. Follow the voice prompt, do one of the following:
 - Press (#) on the phone where the call is parked.
 - Enter the desired extension followed by # (e.g., 4605#) on any phone.

Transfer Mode

To configure call park feature in transfer mode via web user interface:

- 1. Click on Features->Call Pickup.
- 2. Select Transfer from the pull-down list of Call Park Mode.
- 3. Select Enabled from the pull-down list of Call Park.

If **Enabled** is selected, the **Park** soft key will display on the LCD screen during a call, and the **Retrieve** soft key will display on the dialing screen.

4. (Optional.) Enter the call park code in the Call Park Code field.

It is configured for the **Park** soft key. This call park code will also apply to the call park key.

5. (Optional.) Enter the park retrieve code in the Park Retrieve Code field.

It is configured for the **Retrieve** soft key. This park retrieve code will also apply to the retrieve key.

ealink 1466	Status Account Netwo	ork DSSKey	Features Sett	ings Directory Security
Forward&DND	Call Pickup			NOTE
	Directed Call Pickup	Enabled	• 🕜	
General Information	Directed Call Pickup Code	*97	0	Directed Call Pickup Picks up an incoming call on a
A	Group Call Pickup	Enabled	- 0	specific extension.
Audio	Group Call Pickup Code	*98		Directed Call Pickup Picks up incoming calls within
Intercom	Visual Alert for BLF Pickup	Enabled	- 0	pre-defined group.
Transfer				You can configure
	Audio Alert for BLF Pickup	Enabled	• 🕜	directed/group call pickup feature for the IP phone.
Call Pickup	Ring Type for BLF Pickup	Splash.wav	•	Visual Alert for BLF Pickup
Remote Control	Call Park 🕜			It allows the supervisor's pho to display a visual prompt wh
Diama ta di	Call Park Mode	Transfer	• 0	the monitored user receives incoming call.
Phone Lock	Call Park	Enabled	- ()	-
ACD	Call Park Code	*01	0	Audio Alert for BLF Picku It allows the supervisor's pho
SMS	Park Retrieve Code	*11	0	to play an alert tone when t monitored user receives an incoming call.

6. Click **Confirm** to accept the change.

Note If the Park or Retrieve soft key doesn't appear on the LCD screen, please select Disabled from the pull-down list of Custom Softkey via web user interface at path Settings->Softkey Layout.

To park a call in Transfer mode:

 During a call, press the **Park** soft key (You may need to press the **More** soft key to see the **Park** soft key) or the call park key (if configured). - If the call park code is not configured, you need to enter the call park code (e.g., *01).

📞 Park to			
<u>& 101</u> *	01		
Park	123	Delete	Cancel

Press (\mathbf{w}) , $(\#_{\mathbf{w}})$ or the **Park** soft key. The call will be transferred to the shared parking lot.

- If the call park code is configured, the call will be directly transferred to the shared parking lot.
- **Note** For some servers, the system will return a specific call park retrieve number (park retrieve code) from which the call can be retrieved after parking successfully.

To retrieve a parked call in Transfer mode:

- **1.** Do one of the following:
 - If the park retrieve code is not configured, dial the park retrieve code (e.g., *11).



 If the park retrieve code is configured, press the **Retrieve** soft key on the dialing screen or retrieve key (if configured) on the IP phone.

The phone will retrieve the parked call from the shared parking lot.

Call Pickup

You can use call pickup to answer someone else's incoming call on your phone. The SIP-T46G IP phone supports directed call pickup and group call pickup. Directed call pickup is used for picking up a call that is ringing at a target phone number. Group call pickup is used for picking

up a call that is ringing at any phone number in a certain group. The pickup group should be predefined, contact your system administrator for more information.

You can pick up an incoming call by using the **DPickup/GPickup** soft key. To use call pickup, you need to configure the call pickup code beforehand on a global or per-line basis via web user interface.

Note If there are many incoming calls at the same time, pressing the **GPickup** soft key on the phone will pick up the call that rings first.

Directed Call Pickup

To enable directed call pickup and configure the directed call pickup code on a global basis via web user interface:

- 1. Click on Features->Call Pickup.
- 2. Select Enabled from the pull-down list of Directed Call Pickup.
- 3. Enter the directed call pickup code in the **Directed Call Pickup Code** field.

ealink 1466								Log O English(English)
	Status	Account	Network	DSSKey	Featur	es	Settings	Directory Security
Forward&DND	Ca	all Pickup						NOTE
General		Directed Call Pick	up	Enabled	•	0		Dissected Coll Distance
Information		Directed Call Pick	up Code	*97		0		Directed Call Pickup Picks up an incoming call on a
Audio		Group Call Pickup		Enabled	•	0		specific extension.
		Group Call Pickup	Code	*98		0		Directed Call Pickup Picks up incoming calls within a
Intercom		Visual Alert for BL	_F Pickup	Enabled	•	0		pre-defined group.
Transfer		Audio Alert for Bl	LF Pickup	Enabled	•	0		You can configure directed/group call pickup
Call Pickup		Ring Type for BL	F Pickup	Splash.wav	-			feature for the IP phone.
Remote Control	Ca	all Park 🕜						Visual Alert for BLF Pickup It allows the supervisor's phon to display a visual prompt when
Phone Lock		Call Park Mode		Transfer	•	?		the monitored user receives an incoming call.
PHONE LOCK		Call Park		Disabled	•	0		Audio Alert for BLF Pickup
ACD		Call Park Code				0		It allows the supervisor's phone to play an alert tone when the
SMS		Park Retrieve Coo	de			0		monitored user receives an incoming call.
Action URL		Confi	irm		Cancel			You can click here to get
mit is all								more guides.

4. Click **Confirm** to accept the change.

To configure the directed call pickup code on a per-line basis via web user interface:

- 1. Click on Account->Advanced.
- 2. Select the desired account from the pull-down list of Account.

3. Enter the directed call pickup code in the Directed Call Pickup Code field.

ealink 146g				Log Out English(English) -
	Status Account Network	C DSSKey Featu	Ires Settings	Directory Security
Register	Account	Account 1	0	NOTE
Basic	Keep Alive Type	Default 👻	0	DTMF
Dasic	Keep Alive Interval(Seconds)	30	0	It is the signal sent from the IP
Codec	RPort	Disabled 👻	0	phone to the network, which is generated when pressing the IF
Advanced	Subscribe Period(Seconds)	1800	0	phone's keypad during a call.
		÷		Session Timer It allows a periodic refresh of SIP sessions through a re-INVITE request, to determine whether a SIP
	SIP Server Type	Default 👻	0	session is still active.
	Music Server URI	sip:moh@sip.com	0	Busy Lamp Field/BLF List
	Directed Call Pickup Code	*97	0	Monitors a specific extension/a list of extensions for status
	Group Call Pickup Code	*98	0	changes on IP phones.
	Distinctive Ring Tones	Enabled 👻	0	Shared Call Appearance
	Unregister When Reboot	Disabled 👻	0	(SCA)/ Bridge Line Appearance (BLA)
	Out Dialog BLF	Disabled 👻	0	It allows users to share a SIP line on several IP phones. Any
	VQ RTCP-XR Collector name		0	IP phone can be used to originate or receive calls on the
	VO RTCP-XR Collector address		0	shared line.
	VQ RTCP-XR Collector port	5060	0	Network Conference It allows multiple participants
	Confirm	Cancel		(more than three) to join in a call.

4. Click **Confirm** to accept the change.

To pick up a call directly:

1. Pick up the handset, press the Speakerphone key or press the line key (You may need to press the **More** soft key to see the **DPickup** soft key).

The **DPickup** soft key appears on the LCD screen.

% 1008		
<u>§</u> 1008		
1007		
Line	DPickup	More

- 2. Press the **DPickup** soft key on your phone when the target phone receives an incoming call.
- 3. Enter the phone number which is receiving an incoming call.
- 4. Press the **DPickup** soft key again.

The call is answered on your phone.

You can also configure a DSS key as a direct pickup key via phone user interface or web user interface. Once configured, you can pick up a call by pressing the direct pickup key directly. For more information, refer to DSS Keys on page 90.

Group Call Pickup

To enable group call pickup and configure the group call pickup code on a global basis via web user interface:

- 1. Click on Features->Call Pickup.
- 2. Select Enabled from the pull-down list of Group Call Pickup.
- 3. Enter the group call pickup code in the Group Call Pickup Code field.

ealink 1746g						English(English)
	Status Acco	unt Network	DSSKey	Features	Settings	Directory Security
Forward&DND	Call Pickup					NOTE
0	Directed	Call Pickup	Enabled	- 🕜		
General Information	Directed	Call Pickup Code	*97	0		Directed Call Pickup Picks up an incoming call on a
Audio	Group Ca	II Pickup	Enabled	- Q		specific extension.
	Group Ca	I Pickup Code	*98	0		Directed Call Pickup Picks up incoming calls within a
Intercom	Visual Ale	rt for BLF Pickup	Enabled	• 0		pre-defined group.
Transfer	Audio Ale	ert for BLF Pickup	Enabled	• 0		You can configure directed/group call pickup
Call Pickup	Ring Type	e for BLF Pickup	Splash.wav	-		feature for the IP phone.
Remote Control	Call Park	0				Visual Alert for BLF Pickup It allows the supervisor's phon to display a visual prompt whe
Phone Lock	Call Park I	Mode	Transfer	- 0		the monitored user receives a incoming call.
Phone Lock	Call Park		Disabled	- 🕜		Audio Alert for BLF Pickup
ACD	Call Park (Code		0		It allows the supervisor's phon to play an alert tone when the
SMS	Park Retr	ieve Code		0		monitored user receives an incoming call.
Action URL		Confirm		Cancel		You can click here to get
Blustooth			_			more guides.

4. Click **Confirm** to accept the change.

To configure the group call pickup code on a per-line basis via web user interface:

- 1. Click on Account->Advanced.
- 2. Select the desired account from the pull-down list of Account.

3. Enter the group call pickup code in the Group Call Pickup Code field.

Yealink 1466	Status Account Network	DSSKey Featur	res Settings	Log Out English(English) • Directory Security
Register	Account	Account 1 👻	0	NOTE
	Keep Alive Type	Default 👻	0	
Basic	Keep Alive Interval(Seconds)	30	0	DTMF It is the signal sent from the IP
Codec	RPort	Disabled 👻	0	phone to the network, which is generated when pressing the IP
Advanced	Subscribe Period(Seconds)	1800	0	phone's keypad during a call.
		÷		Session Timer It allows a periodic refresh of SIP sessions through a re-INVITE request, to determine whether a SIP
	SIP Server Type	Default 👻	0	session is still active.
	Music Server URI	sip:moh@sip.com	0	Busy Lamp Field/BLF List
	Directed Call Pickup Code	*97	0	Monitors a specific extension/a list of extensions for status
	Group Call Pickup Code	*98	0	changes on IP phones.
	Distinctive Ring Tones	Enabled 👻	0	Shared Call Appearance
	Unregister When Reboot	Disabled 👻	0	(SCA)/ Bridge Line Appearance (BLA)
	Out Dialog BLF	Disabled 👻	0	It allows users to share a SIP line on several IP phones. Any
	VQ RTCP-XR Collector name		0	IP phone can be used to originate or receive calls on the
	VQ RTCP-XR Collector address		0	shared line.
	VQ RTCP-XR Collector port	5060	0	Network Conference It allows multiple participants (more than three) to join in a
	Confirm	Cancel		call.

4. Click **Confirm** to accept the change.

To pick up a call in the group:

1. Pick up the handset, press the Speakerphone key or press the line key.

The **GPickup** soft key appears on the LCD screen.

% 1006							
<u>§</u> 1006]				
1007	1020		Ī				
	1028						
	6026						
	6027						
Directory	123	GPickup	Line				

2. Press the **GPickup** soft key on your phone when a phone in the group receives an incoming call.

The call is answered on your phone.

You can also configure a DSS key as a group pickup key via phone user interface or web user interface. Once configured, you can pick up a call by pressing the group pickup key directly. For more information, refer to DSS Keys on page 90.

Note The direct call pickup code and group call pickup code are predefined on the system server. Contact your system administrator for more information.

The call pickup code configured on a per-line basis takes precedence over that configured on a global basis.

Anonymous Call

You can use anonymous call to block your identify and phone number from appearing to the called party when you call someone. For example, you want to call to consult some of the services, but don't want to be harassed. Anonymous call is configurable on a per-line basis. You can also configure the phone to send anonymous call on/off code to the server to activate/deactivate anonymous call on the server side.

Note Anonymous call is not available on all servers. Contact your system administrator for the anonymous call on code and off code.

To configure anonymous call via phone user interface:

- 1. Press Menu->Features->Anonymous.
- 2. Select the desired line and then press the Enter soft key.
- **3.** Press (•) or (•) , or the **Switch** soft key to select **Enabled** from the **Local Anonymous** field.
- **4.** (Optional.) Press (•) or (•) , or the **Switch** soft key to select the desired value from the **Send Anonymous Code** field.

The phone will send the configured on code or off code depending on your selection when you enable or disable anonymous call feature on the phone.

 (Optional.) Enter the anonymous call on code and off code respectively in the **On Code** and **Off Code** field.

Line 1						
1. Local Anonymous:		Enabled	< ►			
2. Anonymous Rejection:		Disabled	< ►			
3. Send Anonymous Code:		Off Code	∢ ►			
4. On Code:						
5. Off Code:						
Back	123	Delete	Save			

6. Press the Save soft key to accept the change or the Back soft key to cancel.

Anonymous call is configurable via web user interface at the path Account->Basic.

To place an anonymous call:

1. Using the specific line on the phone to place a call to phone B.

The LCD screen of phone B prompts an incoming call from anonymity.



Anonymous Call Rejection

You can use anonymous call rejection to reject incoming calls from anonymous callers. Anonymous call rejection automatically rejects incoming calls from callers who deliberately block their identities and numbers from being displayed. Anonymous call rejection is configurable on a per-line basis. You can also configure the phone to send anonymous call rejection on/off code to the server to activate/deactivate anonymous call rejection on the server side.

To configure anonymous call rejection via phone user interface:

- 1. Press Menu->Features->Anonymous.
- 2. Select the desired line and then press the Enter soft key.
- **3.** Press (•) or (•) to scroll to the **Anonymous Rejection** field.
- Press (•) or (•) , or the Switch soft key to select Enabled from the Anonymous Rejection field.
- 5. (Optional.) Press (•) or (•), or the Switch soft key to select the desired value from the Send Rejection Code field.

The phone will send the configured on code or off code depending on your selection when you enable or disable anonymous call rejection feature on the phone.

 (Optional.) Enter the anonymous call rejection on code and off code respectively in the On Code and Off Code field.

Line 1						
6. Send Rejection Code:		On Code	$\triangleleft \triangleright$			
7. On Code:		1				
8. Off Code:						
Back	123	Delete	Save			

7. Press the **Save** soft key to accept the change or the **Back** soft key to cancel.

Anonymous call rejection is configurable via web user interface at the path **Account->Basic**.

Advanced Phone Features

This chapter provides operating instructions for the advanced features of the SIP-T46G IP phone. Topics include:

- Busy Lamp Field (BLF)
- BLF List
- Call Recording
- Hot Desking
- Intercom
- Multicast Paging
- Music on Hold
- Automatic Call Distribution (ACD)
- Shared Call Appearance (SCA)
- Bridged Lines Appearance (BLA)
- Messages

If you require additional information or assistance with your new phone, contact your system administrator.

Busy Lamp Field (BLF)

You can use BLF to monitor a specific line for status changes on the phone. For example, you can configure a BLF key on the phone to monitor the status of a friend's line (busy or idle). The BLF key LED illuminates solid red when the friend's line is in use. For more information on BLF key LED indicators, refer to LED Instructions on page 9.

Configuring the BLF Feature on the IP Phone

You need to configure the IP phone to perform BLF monitoring. IP phones being monitored do not require any configuration.

Configuring a BLF Key

To configure a BLF key via phone user interface:

- 1. Press Menu->Features->DSS Keys.
- 2. Select the desired DSS key.
- **3.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select **BLF** from the **Type** field.

- **4.** Press () or () , or the **Switch** soft key to select the desired line from the **Account ID** field.
- 5. (Optional.) Enter the string that will appear on the LCD screen in the Label field.
- 6. Enter the phone number or extension you want to monitor in the Value field.
- 7. (Optional.) Enter the directed call pickup code in the Extension field.

If it is configured, when the monitored user receives an incoming call, the supervisor can press the BLF key to pick up the call directly.

Line Key 3						
1. Туре:		BLF	<►			
2. Account ID:		Line 1	<►			
3. Label:		1000				
4. Value:		1000				
5. Extension:		*33				
Back	2aB	Delete	Save			

8. Press the Save soft key to accept the change or the Back soft key to cancel.

BLF key is configurable via web user interface at the path Dsskey->Line Key.

Configuring Visual and Audio Alert for BLF pickup

You can enable visual alert feature for BLF pickup on the phone. This allows the LCD screen of the monitoring phone to display the caller ID when the monitored line receives an incoming call. You can also enable audio alert feature for BLF pickup on the phone. If both of them are enabled, the monitoring phone will play a specific ring tone (refer to BLF Ring Type) when the monitored line receives an incoming call.

To enable visual and audio alert features via web user interface:

- 1. Click on Features->Call Pickup.
- 2. Select Enabled from the pull-down list of Visual Alert for BLF Pickup.

Status Account Network DSSKey Features Settings Directory S Forward&DND Call Pickup Enabled 0	ealink T46G				English(English)
Forward&DND Directed Call Pickup Enabled Image: Control control General Information Directed Call Pickup Enabled Image: Control control control Audio Group Call Pickup Enabled Image: Control c		Status Account Netwo	rk DSSKey	Features Setti	ings Directory Security
General Information Directed Call Pickup Code *97 Image: Composition of the state of	Forward&DND	Call Pickup			NOTE
Information Directed Call Pickup Code *97 Image: Control Call Pickup Code *97 Image: Control Call Pickup Code Pick up an norming specific extension. Audio Group Call Pickup Code *98 Image: Control Call Pickup Code *98 Image: Control Call Pickup Code Pick up an norming specific extension. Transfer Vsual Alert for BLF Pickup Enabled Image: Control Call Pickup Code *98 Image: Control Call Pickup Code You can configure directed/group call pre-defined group. Call Pickup Rng Type for BLF Pickup Enabled Image: Control Call Park FAC Image: Control Call Park Image: Contro	01	Directed Call Pickup	Enabled	▼ (2)	
Audio Group Call Pickup Enabled Image: Call Pickup Intercom Group Call Pickup Code *98 Image: Call Pickup Transfer Visual Alert for BLF Pickup Enabled Image: Call Pickup Audio Alert for BLF Pickup Enabled Image: Call Pickup Call Pickup Remote Control Call Park Image: Call Pickup Phone Lock Call Park Call Park Image: Call Pickup ACD Call Park Code *68 Image: Call Pickup SMS Park Retrieve Code *788 Image: Call Pickup		Directed Call Pickup Code	*97	0	Picks up an incoming call on a
Intercom Group Call Pickup Code "98 ? Pickup uncoming call vickup norming call vickup norming call vickup norming call vickup norming call vickup Pickup norming call vickup Pickup norming call vickup norming norming norming norming norming norming call vickup norming call vickup norming nor		Group Call Pickup	Enabled	- 0	specific extension.
Intercom Visual Alert for BLF Pickup Enabled Image: Control pre-defined group. Call Pickup Remote Control Remote Control Call Park Call Pickup Call Park Call Park Image: Control call park ACD Call Park Code *68 Image: Control call park code	Audio	Group Coll Pickup Code	*09		Directed Call Pickup Picks up incoming calls within
Transfer Audio Alert for BLF Pickup Enabled Vou can configure directed/group call feature for the IP p Call Pickup Ring Type for BLF Pickup Splash.wav Visual Alert for BL t alows the superv to display a visual p Remote Control Call Park FAC Visual Alert for BL t alows the superv to display a visual p Phone Lock Call Park Enabled Visual Alert for BL t alows the superv to display a visual p ACD Call Park Code *68 Visual Alert for BL t alows the superv to play an alert tone montored user recommendation SMS Park Retrieve Code *88 Visual Alert for BL	Intercom				
Call Pickup Range Type for BLF Pickup Splash.wav Visual Alert for BL Remote Control Call Park It alows the supervised of the provided user in coming call. Phone Lock Call Park FAC It alows the supervised user in coming call. ACD Call Park Code F68 It alows the supervised user in coming call. SMS Park Retrieve Code F88 It alows the supervised user in coming call.	Transfor				
Call Pickup Ring Type for BLF Pickup Splash.wav Visual Alert for BL It alows the superv to display a visual price and price control Remote Control Call Park @ FAC ? Phone Lock Call Park Mode FAC ? ACD Call Park Code *68 ? SMS Park Retrieve Code *88 ?	Tansici	Audio Alert for BLF Pickup	Enabled	- ()	directed/group call pickup feature for the IP phone.
Remote Control Call Park It alows the supervised of the monitored user in coming call Phone Lock Call Park Mode FAC It alows the supervised of the monitored user in coming call ACD Call Park Code F68 It alows the supervised of the monitored user in coming call SMS Park Retrieve Code F88 It alows the supervised of the monitored user in coming call	Call Pickup	Ring Type for BLF Pickup	Splash.wav	-	Visual Alert for BLF Pickup
Phone Lock Cal Park Mode FAC Image: Cal Park Mode Cal Park Cal Park Enabled Image: Cal Park Mode ACD Cal Park Code *68 Image: Cal Park Mode SMS Park Retrieve Code *88 Image: Cal Park Mode	Remote Control	Call Park 🕜			It allows the supervisor's phot to display a visual prompt whe
Call Park Enabled Image: Coll Park ACD Call Park Code *68 Image: Coll Park Code SMS Park Retrieve Code *88 Image: Coll Park Code	Diama ta di	Call Park Mode	FAC	- 0	the monitored user receives a
ACD Cal Park Code *68 ? It alows the superv to play an alert tone SMS Park Retrieve Code *88 ? monitored user rece	Phone Lock	Call Park	Enabled	- ()	-
SMS Park Retrieve Code *88 2 monitored user rece	ACD	Call Park Code	*68	0	It allows the supervisor's pho
incoming call.	SMS	Park Retrieve Code	*88		to play an alert tone when th monitored user receives an incoming call.
Action URL Confirm Cancel	Action URL	Confirm		Cancel	You can click here to get

3. Select Enabled from the pull-down list of Audio Alert for BLF Pickup.

4. Click **Confirm** to accept the change.

Note Visual and audio alert features are configurable via web user interface only.

When the monitored line receives an incoming call, the following occurs on your phone:

- The phone plays a specific ring tone (if enabled).
- The BLF key LED flashes red.
- The caller ID appears on the LCD screen (if enabled).

In the following figure, the LCD screen shows an incoming call from 1008 on the monitored line.



You can do the following actions:

- Press the **DPickup** soft key to pick up the call.
- Press the **Send** soft key to dial out the monitored phone number.
- Press the **New Call** soft key to place a new call.

- Press the Cancel soft key to exit the screen.
- Note If your phone is locked and the type of the phone lock is configured as **Function Keys** or **All Keys**, you cannot use the **DPickup**, **Send**, **New Call** and **Cancel** soft keys until unlocked. For more information refer to Phone Lock on page 53.

Configuring Directed Call Pickup Code

If you configure the directed call pickup code in advanced, you can pick up the monitored user's incoming call by pressing the BLF key. You can configure the directed call pickup code when configuring a BLF key. For more information, refer to Configuring a BLF Key on page 163. You can also configure the directed call pickup code on a global or per-line basis via web user interface.

If you enable visual alert feature, you can also pick up the monitored user's incoming call by pressing **DPickup** soft key. To use **DPickup** soft key to pick up the call, the call pickup code can only be configured on a global or per-line basis via web user interface.

Note The pickup code is used in the following order of preference: Extension (Dsskey->Line Key)>Directed Call Pickup Code (Account->Advanced)>Directed Call Pickup Code (Features->Call Pickup). If pickup code is not configured, pressing the BLF key/DPickup soft key will directly call the monitored user when he/she receives an incoming call.

To configure the directed call pickup code on a global basis via web user interface:

- 1. Click on Features->Call Pickup.
- 2. Enter the desired pickup code in the **Directed Call Pickup Code** field.

Yealink 1465							Eng	Log Out ish(English) 🗸
	Status Account	Network	DSSKey	Feature	es	Settings	Directory	Security
Forward&DND	Call Pickup						NOTE	
	Directed Call Pickup		Enabled	•	?			
General Information	Directed Call Pickup	Code	*97		0		Directed Call Picks up an inc	oming call on a
Audio	Group Call Pickup	[Enabled	•	0		specific extens	
Audio	Group Call Pickup C	ode	*98	_	0		Directed Call Picks up incom	Pickup ng calls within a
Intercom	Visual Alert for BLF	Pickup	Enabled	•	0		pre-defined gro	oup.
Transfer	Audio Alert for BLF		Enabled	•	0		You can config directed/group	
Call Pickup	Ring Type for BLF F		Ring1.wav	•			feature for the	IP phone.
Remote Control	Call Park 🕜	ickup	Ning1.wav	·				or BLF Pickup pervisor's phone al prompt when
Phone Lock	Call Park Mode		Transfer	-	0			user receives an
ACD	Call Park Call Park Code		Enabled	•	0		Audio Alert fo	or BLF Pickup
SMS	Call Park Code Park Retrieve Code		*11		0 0			tone when the
Action URL	Confirm	n		Cancel			🛽 You can cli	ck here to get

3. Click **Confirm** to accept the change.

To configure the directed call pickup code on a per-line basis via web user interface:

1. Click on Account->Advanced.

2. Select the desired account from the pull-down list of Account.

3.	Enter the desired pickup code in the Directed Call Pickup Code field.

Yealink	Status Account Network	DSSKey Features Setting	Log Out English(English) - Directory Security
Register	Account	Account 1 • ?	NOTE
Basic	Keep Alive Type	Default 🗸 🕜	DTMF
Codec	Keep Alive Interval(Seconds) RPort	30 🕜 Disabled 👻 🕜	It is the signal sent from the IP phone to the network, which is generated when pressing the IP
Advanced	Subscribe Period(Seconds)	1800	phone's keypad during a call.
		:	Session Timer It allows a periodic refresh of SIP sessions through a re-INVITE request, to determine whether a SIP
	SIP Server Type	Default 🗸 🥜	session is still active.
	Music Server URI	sip:moh@sip.com	Busy Lamp Field/BLF List Monitors a specific extension/a
	Directed Call Pickup Code Group Call Pickup Code	*97 2	list of extensions for status changes on IP phones.
	Distinctive Ring Tones	Enabled 🗸 🖓	Shared Call Appearance
	Unregister When Reboot	Disabled 🗸 🕜	(SCA)/ Bridge Line Appearance (BLA) It allows users to share a SIP
	Out Dialog BLF	Disabled 🗸 🕜	line on several IP phones. Any IP phone can be used to
	VQ RTCP-XR Collector name VO RTCP-XR Collector address		originate or receive calls on the shared line.
	VQ RTCP-XR Collector port	5060 2	Network Conference It allows multiple participants
	Confirm	Cancel	(more than three) to join in a call.

4. Click **Confirm** to accept the change.

Configuring Transfer Mode on DSS Key

When there is an active call on the IP phone, you can handle the active call differently depending on the transfer mode on DSS key.

To configure the transfer mode on DSS key via web user interface:

- 1. Click on Features->Transfer.
- 2. Select the desired transfer mode from the pull-down list of Transfer Mode via Dsskey.
 - If you select **Blind Transfer**, press the BLF key to blind transfer the call to the monitored user.
 - If you select **Attended Transfer**, press the BLF key to dial out the monitored phone number and then perform an attended or a semi-attended transfer.
 - If you select **New Call**, press the BLF key to place a new call to the monitored user.

Yealink 1466	Status Account Network	DSSKey Features	Log Out English/(English) V Settings Directory Security
Forward&DND	Transfer		NOTE
General	Semi-Attended Transfer	Enabled 🗸 🥜	a # 7
Information	Blind Transfer on Hook	Enabled 🗸 🥎	Call Transfer The transfer parameters for
Audio	Attended Transfer on Hook	Enabled V	administrator.Call transfer enables IP phones to transfer an existing call to another party. IP
Intercom	Transfer on Conference Hang up	Disabled 🗸 🥜	phones support call transfer using the REFER method
Intercom	Transfer Mode via Dsskey	Blind Transfer 🗸 🥜	specified in RFC 3515 and offer
Transfer			three types of transfer.
Call Pickup	Confirm	Cancel	Blind Transfer Transfer a call directly to another

3. Click **Confirm** to accept the change.

Using BLF Feature on the IP Phone

You can use the BLF key in the following ways:

- Place a call to the monitored user.
- Transfer a call to the monitored user.
- Pick up an incoming call of the monitored user.

Placing a Call to the Monitored User

You can press a BLF key to dial out the monitored phone number when the monitored line is idle.

To place a call to the monitored user when the IP phone is idle:

1. Press the BLF key.

The IP phone dials out the phone number of the monitored user.

Transferring a Call to the Monitored User

When there is already an active call on the IP phone, you can transfer the active call to the monitored user by pressing the BLF key. The phone handles the active call differently depending on the transfer mode on DSS key (refer to Configuring Transfer Mode on DSS Key). The transfer mode on DSS key has the following modes:

- Blind Transfer
- Attended Transfer
- New Call

When the transfer mode on DSS key is set to **Blind Transfer** and there is already an active call on the IP phone, you can do the following:

To perform a blind transfer:

1. Press the BLF key of the monitored user which you want to transfer this call to.

Then the active call is connected to the number of the monitored user.

When the transfer mode on DSS key is set to **Attended Transfer** and there is already an active call on the IP phone, you can do the following:

To perform a semi-attended transfer:

- **1.** Press the BLF key of the monitored user which you want to transfer this call to. The phone will dial the number of the monitored user.
- **2.** Press the BLF key of the monitored user again or the **Transfer** soft key to complete the transfer when receiving ringback.

To perform an attended transfer:

- **1.** Press the BLF key of the monitored user which you want to transfer this call to. The phone will dial the number of the monitored user.
- **2.** Press the BLF key of the monitored user again or the **Transfer** soft key to complete the transfer after the monitored user answers the call.

When the transfer mode on DSS key is set to **New Call** and there is already an active call on the IP phone, you can do the following:

To perform a new call:

1. Press the BLF key of the monitored user you want to call.

The phone will dial the number of the monitored user.

Picking up an Incoming Call of the Monitored User

When the monitored user receives an incoming call, the BLF key indicator of the monitored user will flash red. If you enable visual and audio alert features on the phone, the phone will play a specific ring tone (refer to BLF Ring Type) and display the incoming call information on the screen. You can use the BLF key to pick up an incoming call of the monitored user. Before picking up an incoming call, ensure that the directed call pickup code has been configured in advance.

To pick up an incoming call when the IP phone is idle:

1. Press the BLF key.

The incoming call of the monitored user is answered on the IP phone.

Note If the call pickup code is not configured, the phone will place a call to the monitored user instead of picking up an incoming call of the monitored user when you press the BLF key.

BLF List

You can use the BLF List feature to monitor a list of users defined by your system administrator. For example, your system administrator enables BLF List, and creates a BLF List URI (e.g., BLFList@example.com) including a list of user1, user2 on the server. The BLF List keys on the IP phone can present the status of user1 and user2. The key LEDs illuminate either flashing or solid depending on the status of those users. For more BLF List key LED indicators, refer to LED Instructions on page 9.

You can use the BLF List keys in the following ways:

- When the monitored user is idle, press the BLF list key to dial out the phone number.
- When there is already an active call on the IP phone, you can transfer the active call to the monitored user by pressing the BLF List key. The phone handles the active call differently depending on the transfer mode on DSS key. For more information on the transfer mode

on DSS key, refer to Configuring Transfer Mode on DSS Key on page 166.

- When the monitored user receives an incoming call, press the BLF list key to pick up the call directly. Before picking up an incoming call, ensure that the BLF List Pickup Code has been configured in advance. If the code is not configured, the phone will place a call to the monitored user instead of picking up the incoming call of the monitored user when you press the BLF List key.
- When there is a conversation on the monitored user, press the BLF list key to barge in and set up a conference call. Before barging in an active call, ensure that the BLF List Barge In Code has been configured in advance. If the code is not configured, the phone will place a call to the monitored user instead of barging in an active call of the monitored user when you press the BLF List key.
- When a call is being parked against the monitored phone, press the BLF List key to retrieve the parked call from the monitored user. Before retrieving the parked call, ensure that the BLF List Retrieve Call Parked Code has been configured in advance. If the code is not configured, the phone will place a call to the monitored user instead of retrieving the parked call when you press the BLF List key.

To configure BLF List settings via web user interface:

- 1. Click on Account->Advanced.
- 2. Select the desired account from the pull-down list of Account.
- 3. Enter the BLF List URI in the BLF List URI field.
- 4. (Optional.) Enter the directed call pickup code in the BLF List Pickup Code field.
- 5. (Optional.) Enter the barge-in code in the **BLF List Barge In Code** field.
- 6. (Optional.) Enter the park retrieve code in the BLF List Retrieve Call Parked Code field.

ealink 146G								Eng	Log lish(English)
	Status	Account	Network	DSSKey	Featu	res	Settings	Directory	Security
Register	Acc	ount		Account 1	Ŧ	?		NOTE	
Basic	Kee	p Alive Type		Default	-	0		DTHE	
Basic	Kee	p Alive Interval(See	conds)	30		0		DTMF It is the signal	
Codec	RPor	t		Disabled	-	0		phone to the generated who	en pressing th
Advanced						-		phone's keypa	d during a call.
				20		0		Session Time It allows a peri SIP sessions th re-INVITE requ	odic refresh of rough a
		ne(ms)			•	<u> </u>		determine whether a SIP session is still active.	ether a SIP
	BLF	List URI		4609@pbx.yealir	nk.com	0		Session is suil a	cuve.
	BLF	List Pickup Code		*97		0		Busy Lamp Fi	
	BLF	List Barge In Code		*33		0		Monitors a spe- list of extension	ns for status
	BLF	List Retrieve Call Pa	arked Code	*88		0		changes on IP	phones.
	Shar	ed Line		Shared Call Appe	arance 👻	0		Shared Call A	ppearance
	Call	Pull Feature Access	Code	*11		0		(SCA)/ Bridge Appearance (e Line
	Dialo	g Info Call Pickup		Disabled	-	0		It allows users to share a line on several IP phones.	to share a SIP
	VOI	RTCP-XR Collector	name			0		IP phone can be used to originate or receive calls o shared line.	
		RTCP-XR Collector				0			
	-					- -			
	VQ	RTCP-XR Collector	port	5060		0		Network Con It allows multip	
		Confi	rm		Cancel]		(more than the	ree) to join in a

- 7. Click **Confirm** to accept the change.
- **Note** For more information on BLF List URI/BLF List Pickup Code/BLF List Barge In Code/BLF List Retrieve Call Parked Code, contact your system administrator.

According to the response message from the server, the IP phone will automatically configure the BLF List keys beginning from the first unused DSS key (The default order of BLF list keys assigned automatically is Line Key->Ext Key). Once any DSS key is seized, the IP phone will skip to configure the next DSS key.

You can receive a visual or/and an audio alert (if enabled) on your phone when the monitored user receives an incoming call. For more information, refer to Configuring Visual and Audio Alert for BLF pickup on page 164.

Note

The pickup code is used in the following order of preference: BLF List Pickup Code (**Account->Advanced**)>Directed Call Pickup Code (**Account->Advanced**)>Directed Call Pickup Code (**Features->Call Pickup**). If all of them are not configured, pressing the BLF List key will directly call the monitored user when he/she receives an incoming call. For more information on pickup code, refer to Call Pickup on page 154.

Call Recording

The SIP-T46G IP phone supports local and server recording. If you insert a USB flash drive into the USB port on your phone, you can record active calls on your phone. The recordings are saved as *.wav files on your USB flash drive. You can also record active calls by pressing a record key on the phone. The recordings are saved on the server.

Note Before recording any call, especially those involving PSTN, it is necessary to know about the rules and restrictions of any governing call-recording in the place you are in. It is also very important to have the consent of the person you are calling before recording the conversation.

Recording Using a USB Flash Drive

You can attach a USB flash drive to your IP phone when the phone is idle or during an active call. This feature is not available by default. For more information, contact your system administrator.

When the phone is idle, and you insert a USB flash drive into the USB port on IP phone, the phone will detect the USB flash drive and display a flashing icon 📋 . After detecting the USB flash drive, the icon stops flashing and LCD screen prompts "USB device has been connected successfully!".



When there is an active call, and you insert a USB flash drive into the USB port on your phone, the LCD screen will prompt "USB device has been connected successfully! Available Recording Time: XX:XX:XX". For more information on the available recording time, refer to Viewing the Storage Space on page 177.

The following figure shows an example:

🔹) Talking)		HD 🕲
% 1025	1026		1
0	Note		
US		n connected succe rding Time:01:22:	
Transfe	er Hold	Conference	More

The Start REC soft key controls the recording function, and is available:

- The free space of USB flash drive is sufficient
- When there are one or more calls connected to your IP phone
- During an active call
- When calls are on hold or muted
- During an attended transfer
- During a conference call

The Start REC soft key is not available when:

- You place a new call
- No free space of USB flash drive
- The phone prompts you to answer an incoming call

Call Recording

You can record active calls or conferences.

To record a call:

1. Press the More soft key, and then press the Start REC soft key during a call.

The LCD screen displays the icon (●) and recording duration.



During the recording, you can do the following:

Press the More soft key, and then press the Pause REC soft key. Then (II) appears on the LCD screen and the duration stops counting. The Resume REC soft key appears on the LCD screen. Both you and other parties are not recorded.

Press the **Resume REC** soft key to continue recording.

 Press the More soft key, and then press the Stop REC soft key. The Start REC soft key appears on the LCD screen. Recording stops and the LCD screen prompts "Call has been successfully recorded on USB device".



Press the **Start REC** soft key to start recording again, the recording will be saved in a new ".wav" file.

Press or the Hold soft key while recording, recording is paused. Then II appears on the LCD screen and the duration stops counting. Both you and the callee are not recorded.

Press the **Resume** soft key to resume the call, the recording is continued.

- Press () while recording, the muted party is not recorded.
- If there is insufficient free space (less than 30 minutes) on the USB flash drive during recording, the LCD screen prompts the following:

C T	Falking	● 00 : 19	HD 🍩
% 1	.006	1020	
	🚺 No	ote	
		30 minutes left for call recording	
(Cancel		

Press the Cancel soft key to return to the previous screen.

 If there is insufficient free space (less than 10 minutes) on the USB flash drive during recording, the LCD screen prompts the following:

٤	Talking	● 01 : 18	HD 🛿
S	1006	1020	
	🚺 N	ote	
		10 minutes left for call recording	
	Cancel		

Press the Cancel soft key to return to the previous screen.

If no free space of the USB flash drive during recording, recording is stopped and saved on the USB flash drive automatically. The record soft keys disappear. You cannot record a call unless the space of your USB flash drive is released.

Note You can transfer or set up a conference call while recording. All the calls are recorded in the same way and will be saved in as a ".wav" file.

When you end a call while recording, the recording will be stopped and saved as *.wav files on

the USB flash drive automatically.

To stop recording while the phone records, do one of the following:

- Press the More soft key, and then press the Stop REC soft key.

The LCD screen prompts "Call has been successfully recorded on USB device", the recording icon and recording duration disappear.

📞 Talking						
<u>§</u> 1006	1020					
🚺 N	ote		SALAR SALAR			
Call has been successfully recorded on USB device						
Transfer	Hold	Conference	More			

- Press the More soft key and then press the End Call soft key.

The LCD screen prompts "Call has been successfully recorded on USB device", the recording icon and recording duration disappear, and the phone returns to the idle screen.



Note Unplugging the USB flash drive during a recording call will stop call recording, the recording will be saved on the USB flash drive automatically.

Playing Back Recorded Calls

You can play back recorded calls on your phone. You can also browse and delete the recorded files on the USB flash drive.

Note Playback can occur on either the phone itself or on a computer using an application capable of playing ".wav" files.

To browse the recorded files:

1. Press Menu->USB->Browse Audio.

The LCD screen shows all recorded files on the USB flash drive. The filename consists of the recording time & date, the other party's number/IP address/name (or the first person's number/IP address/name you called).

Browse Auido					
20160701-0954-2025.wav	00:00:01(5	56.9KB)			
20160701-0952-2025(1).wav	00:00:05(1	168.2KB)			
20160701-0952-2025.wav	00:00:05(161.3KB)				
20160701-0951-2025.wav	00:00:30(9	954.4KB)			
20160629-1333-2025.wav	00:00:01(6	51.9KB)			
Back	Delete	Play			

- **2.** Press (\bullet) or (\bullet) to scroll through the recorded files.
- **3.** Press the **Play** soft key.

	20160714-1346-1025.wav					
File Size:	File Size: 779.4KB					
Playing:	Playing: 00:00:06 / 00:00:24					
	o					
Back	Fast Rewind	Fast Forward	Pause			

During the recording playback, you can do the following:

Press the **Pause** soft key to pause the playback. The **Play** soft key appears on the LCD screen.

Press the **Play** soft key to continue the playback.

- Press the Fast Forward soft key to skip forward the playback. Press once to skip forward 8 seconds.
- Press the Fast Rewind soft key to rewind the playback. Press once to rewind 8 seconds.
- Press + to adjust the volume of audio output.
- Press the **Back** soft key to stop the playback. The phone returns to the Browse Audio screen.
- Press the Speakerphone key, the HEADSET key, or by picking up the handset to alternate between Speakerphone, Headset and Handset modes.

If you are playing back a recorded call and an incoming call arrives on your phone, the playback

pauses and the phone rings. The playback will not continue until you press the **Play** soft key.

Deleting the Recorded File

To delete a recorded file:

- 1. Press Menu->USB->Browse Audio.
- 2. Press () or () to highlight the recording you want to delete.
- 3. Press the **Delete** soft key.

The LCD screen prompts the following warning:

	Browse A	udio	
20160713-0	928-104(1).wav	00:00:34(1.0MB)	
2016 1 Not	e		3)
2016	Delete this	audio?	
Cancel		0	к

4. Press the **OK** soft key to confirm the deletion or the **Cancel** soft key to cancel.

Viewing the Storage Space

You can view the storage space (e.g., total space, used space, free space) of the USB flash drive on the phone.

To view the storage space:

1. Press Menu->USB->Storage Space.

5GB
3GB(17.5% used)
3GB(82.5% free)
:24:58

Recording Using a Record Key

There are two ways of call recording in server:

- Record: The phone sends SIP INFO message containing a specific header "Record: on/off" to trigger a recording.
- **URL Record**: The phone sends HTTP URL request to trigger a recording. Contact your system administrator for the predefined URL.
- **Note** Server recording is not available on all servers. Contact your system administrator for more information.

The Record and URL Record keys control the recording function, and are available:

- During an active call
- When calls are on hold or muted
- During a blind or attended transfer
- During a conference call
- When the phone prompts you to answer an incoming call

The Record and URL Record keys are not available when:

- There are not connected calls on your phone
- You place a new call

Configuring a Record Key

To configure a record key via phone user interface:

- 1. Press Menu->Features->DSS Keys.
- 2. Select the desired DSS key.
- **3.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select **Key Event** from the **Type** field.
- **4.** Press (•) or (•) , or the **Switch** soft key to select **Record** from the **Key Type** field.
- 5. (Optional.) Enter the string that will appear on the LCD screen in the Label field.

Line Key 1				
1. Type:	Key Event	$\triangleleft \triangleright$		
2. Key Type:	Record			
3. Label:				
Back	Switch	Save		

6. Press the Save soft key to accept the change or the Back soft key to cancel.

Record key is configurable via web user interface at the path **Dsskey->Line Key**.

Configuring a URL Record Key

To configure a URL record key via phone user interface:

- 1. Press Menu->Features->DSS Keys.
- 2. Select the desired DSS key.
- **3.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select **URL Record** from the **Type** field.
- 4. (Optional.) Enter the string that will appear on the LCD screen in the Label field.
- 5. Enter the URL (e.g., http://10.1.2.224/phonerecording.cgi) in the URL Record field.

	Line	e Key 3	
1. Type: URL Record <			
2. Label:		URL Record	
3. URL Record:		http://10.1.2.224/phone	
Back	abc	Delete	Save

6. Press the Save soft key to accept the change or the Back soft key to cancel.

URL record key is configurable via web user interface at the path Dsskey->Line Key.

Recording a Call

To record a call:

1. Press the record key or URL record key during a call.

If the recording starts successfully, the recording icon will appear on the LCD screen and the record key or URL record key LED will flash green.

2. Press the Record or URL Record key again to stop recording.

The recording icon disappears from the LCD screen and the record key or URL record key LED goes out.

Recording status indicators you need to know:

Circumstance	Icons on the LCD screen	
A recording is started	 appears on the LCD screen 	
A recording cannot be started	X appears for 1 second	
A recording cannot be stopped	ippears for 1 second, then goes back	
The recording box is full	🍿 appears for 1 second	

The call cannot be recorded

🔞 appears for 1 second

You can listen to the recordings saved on your server system. For example, you can dial an access code to listen to the recordings.

Note

The way in which you listen to the recordings may be different depending on the server. Contact your system administrator for more information.

Hot Desking

Hot desking originates from the definition of being the temporary physical occupant of a work station or surface by a particular employee. A primary motivation for hot desking is cost reduction. This feature is regularly used in places where not all the employees are in the office at the same time, or not in the office for very long, which means that actual personal offices would be often vacant, consuming valuable space and resources.

You can use hot desking on the SIP-T46G IP phone to log out of existing accounts and then log into a new account. As a result, many users can share the phone resource at different times. To use this feature, you need to configure a hot desking key in advance.

Note

Hot desking is not available on all servers. Contact your system administrator for more information.

To configure a hot desking key via phone user interface:

- 1. Press Menu->Features->DSS Keys.
- 2. Select the desired DSS key.
- **3.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select **Key Event** from the **Type** field.
- **4.** Press (•) or (•) , or the **Switch** soft key to select **Hot Desking** from the **Key Type** field.
- 5. (Optional.) Enter the string that will appear on the LCD screen in the Label field.

Line Key 1				
1. Type:	Key Event	$\triangleleft \triangleright$		
2. Key Type:	Hot Desking	$\triangleleft \triangleright$		
3. Label:				
S. Lubel.				
Back	Switch	Save		

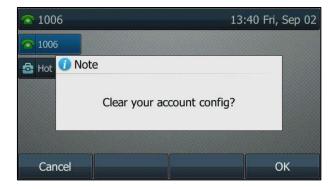
6. Press the Save soft key to accept the change or the Back soft key to cancel.

Hot desking key is configurable via web user interface at the path Dsskey->Line Key.

To use hot desking:

1. Press the hot desking key when the phone is idle.

The LCD screen prompts the following warning:



2. Press the **OK** soft key, registration configurations of all accounts on the phone will be cleared immediately.

The login wizard will be displayed as below:

Hot Desking					
1. User Name:					
2. Password:					
Back	2aB	Delete	Save		

- 3. Enter the login information in each field.
- 4. Press the Save soft key to login or the Back soft key to cancel.

Intercom

Intercom is a useful feature in an office environment to quickly connect with the operator or the secretary. You can press the intercom key to automatically connect with a preconfigured target extension for outgoing intercom calls, and the target extension will automatically answer incoming intercom calls by default. You can also use intercom key to monitor a specific line for status changes on the phone.

Note

Intercom is not available on all servers. Contact your system administrator for more information.

Configuring the Intercom Feature on the IP Phone

Configuring an Intercom Key

To configure an intercom key via phone user interface:

- 1. Press Menu->Features->DSS Keys.
- 2. Select the desired DSS key.
- **3.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select **Intercom** from the **Type** field.
- 4. Select the desired line from the Account ID field.
- 5. (Optional.) Enter the string that will appear on the LCD screen in the Label field.
- 6. Enter the target extension number in the Value field.
- 7. (Optional.) Enter the directed call pickup code in the Extension field.

If it is configured, when the target extension receives an incoming call, the supervisor can press the intercom key to pick up the call directly. Contact your system administrator for more information.

8. Press the Save soft key to accept the change or the Back soft key to cancel.

Line Key 3				
1. Type:		Intercom	$\triangleleft \triangleright$	
2. Account ID	:	Line 1	< ►	
3. Label:				
4. Value: 1000				
5. Extension: *20*				
Back	123	Delete	Save	

Intercom key is configurable via web user interface at the path Dsskey->Line Key.

Incoming Intercom Calls

The SIP-T46G IP phone supports automatically to answer an incoming intercom call by default. The phone automatically plays a warning tone when it receives an incoming intercom call. In addition, you can enable the phone to mute the microphone when it automatically answers an incoming intercom call. You can also enable the phone to automatically answer an incoming intercom call while there is already an active call on the phone. The active call is then placed on hold.

Intercom Feature	Description	
Intercom Allow	Enable or disable the IP phone to automatically answer an	
Intercom Allow	incoming intercom call.	
Intercom Mute	Enable or disable the IP phone's microphone for intercom	

Intercom features you need to know:

Intercom Feature	Description	
	calls.	
Intercom Tone	Enable or disable the IP phone to play a warning tone	
Intercom rone	when it receives an incoming intercom call.	
	Enable or disable the IP phone to automatically answer an	
Intercom Barge	incoming intercom call while there is already an active call	
	on the phone.	

Intercom Allow

You can enable or disable the phone to automatically answer an incoming intercom call. If Intercom Allow is enabled, the phone will automatically answer an incoming intercom call. If Intercom Allow is disabled, the phone will handle an incoming intercom call like a normal call instead of directly rejecting it. Accept Intercom is enabled by default.

Note

Your administrator can set a period of delay time before the phone automatically answers intercom calls. Contact your system administrator for more information.

Intercom Mute

You can mute or un-mute the phone's microphone for intercom calls automatically. If Intercom Mute is enabled, the microphone will be muted for intercom calls. If Intercom Mute is disabled, the microphone will work for intercom calls. Intercom Mute is disabled by default.

Intercom Tone

You can enable or disable the phone to play a warning tone when receiving an intercom call. If Intercom Tone is enabled, the phone will play a warning tone before answering the intercom call. If Intercom Tone is disabled, the phone will automatically answer the intercom call without warning. Intercom Tone is enabled by default.

Intercom Barge

You can enable or disable the phone to automatically answer an incoming intercom call while there is already an active call on the phone. If Intercom Barge is enabled, the phone will automatically answer the intercom call and place the active call on hold. If Intercom Barge is disabled, the phone will handle an incoming intercom call like a waiting call. Intercom Barge is disabled by default.

Note

To enable the phone to receive a new incoming call when it has an active call, make sure that call waiting feature is enabled on the phone in advance. For more information, refer to Call Waiting on page 146.

To configure intercom features via phone user interface:

- 1. Press Menu->Features->Intercom.
- 2. Make the desired changes.

	Intercom	
1. Intercom Allow:	Enabled	$\triangleleft \triangleright$
2. Intercom Mute:	Disabled	$\triangleleft \triangleright$
3. Intercom Tone:	Enabled	∢ ►
4. Intercom Barge:	Disabled	<►
Back	Switch	Save

3. Press the Save soft key to accept the change or the Back soft key to cancel.

Intercom features are configurable via web user interface at the path Features->Intercom.

Using Intercom

You can use the intercom key in the following ways:

- Place a call to the target extension
- Pick up an incoming call of the target extension
- **Note** To use the intercom key to pick up an incoming call, make sure the intercom subscription feature is enabled. For more information, contact your system administrator.

Placing an Intercom Call to the Target Extension

To place an intercom call when the target phone is idle:

1. Press the intercom key when the phone is idle.

The target extension plays a warning tone and automatically answers the call in the speakerphone mode by default.

2. To end the intercom call, press the intercom key again or the End Call soft key.

Picking up an Incoming Call of the Target Extension

When the target extension receives an incoming call, the intercom key LED of the target extension will flash green. If you configure the directed call pickup code in advance, you can pick up the target extension's incoming call by pressing the intercom key. You can configure the directed call pickup code when configuring an intercom key. For more information, refer to Configuring an Intercom Key on page 182.

If you enable visual alert feature, you can also pick up the target extension's incoming call by pressing the **DPickup** soft key. To use the **DPickup** soft key to pick up the call, the call pickup code can only be configured on a global or per-line basis via web user interface. For more information, refer to refer to Configuring Visual and Audio Alert for BLF pickup on page 164 and Configuring Directed Call Pickup Code on page 166.

To pick up an incoming call when the monitored phone is ringing:

1. Press the intercom key.

The incoming call of the target extension is answered on the IP phone.

Note If the directed call pickup code is not configured, the phone will place a call to the target extension instead of picking up an incoming call of the target extension when you press the intercom key.

Multicast Paging

You can use multicast paging to quickly and easily broadcast time sensitive announcements to people who are listening to a specific multicast group. You can configure a multicast paging key or paging list key on the phone, which allows you to send a Real Time Transport Protocol (RTP) stream to the pre-configured multicast address(es) without involving SIP signaling. You can configure the phone to receive an RTP stream from pre-configured multicast listening address(es) without involving SIP signaling. You can specify up to 31 multicast listening addresses.

Yealink IP phone supports the following 31 channels:

- **0**: It is the default channel. This channel is for compatibility with the Yealink IP phones running firmware version 80 or prior and third-party devices.
- **1 to 25**: These channels are for compatibility with the Polycom IP phones.
- 26 to 30: These channels are used for separate communication among the Yealink IP phones running firmware version 81 or later.

The IP phones will automatically ignore all incoming multicast paging calls on the different channel.

Sending RTP Stream

To configure a multicast paging key via phone user interface:

- 1. Press Menu->Features->DSS Keys.
- 2. Select the desired DSS key.
- **3.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select **Key Event** from the **Type** field.
- **4.** Press (•) or (•) , or the **Switch** soft key to select **Multicast Paging** from the **Key Type** field.

- 5. (Optional.) Enter the string that will appear on the LCD screen in the Label field.
- Enter the multicast IP address and port number (e.g., 224.5.6.20:10008) in the Value field.
 The valid multicast IP addresses range from 224.0.0.0 to 239.255.255.255.
- 7. Enter the desired channel in the **Channel** field.

The valid channel ranges from 0 to 30.

Line Key 2				
1. Туре:		Key Event	∢ ►	
2. Key Type:		Multicast Pagin	ig < Þ	
3. Label:				
4. Value: 224.5.6.20:10008			008	
5. Channel:				
Back	123	Delete	Save	

8. Press the Save soft key to accept the change or the Back soft key to cancel.

Multicast paging key is configurable via web user interface at the path Dsskey->Line Key.

To configure a paging list key via phone user interface:

- 1. Press Menu->Features->DSS Keys.
- 2. Select the desired DSS key.
- 3. Press (\bullet) or (\bullet) , or the **Switch** soft key to select **Key Event** from the **Type** field.
- 4. Press (\cdot) or (\cdot) , or the Switch soft key to select Paging List from the Key Type field.
- 5. (Optional.) Enter the string that will appear on the LCD screen in the Label field.

1	Line Key 3					
1. Type:		Key Event	∢ ⊳			
2. Key Type:		Paging List	< ►			
3. Label:		1				
Back	2aB	Delete	Save			

6. Press the Save soft key to accept the change or the Back soft key to cancel.

Paging list key is configurable via web user interface at the path **Dsskey**->Line Key.

To configure paging list via phone user interface:

1. Press the paging list key when the phone is idle.

If the paging list key is not configured, you can also press

Menu->Features->Others->Paging List to configure the paging list.

2. Press (\bullet) or (\bullet) to select a desired paging group.

The default tag is Empty if it is not configured before.

Pa	aging List	
1. (Empty)		
2. (Empty)		
3. (Empty)		
4. (Empty)		
5. (Empty)		
Back	Option	Paging

- 3. Press the Option soft key, and then press the Edit soft key from the prompt list.
- Enter the multicast IP address and port number (e.g., 224.5.6.20:10008) in the Address field. The valid multicast IP addresses range from 224.0.00 to 239.255.255.255.
- 5. Enter the group name in the Label field.
- 6. Enter the desired channel in the **Channel** field.

The valid channel ranges from 0 to 30.

	Edit Pagir	ng Address				
1. Number:		1				
2. Address:		224.5.6.20:100	008			
3. Label:		Product				
4. Channel:		0				
Back	123	Delete	Save			

- 7. Press the **Save** soft key to accept the change.
- 8. Repeat steps 2 to 7, you can add more paging groups.

Paging list is configurable via web user interface at the path Directory->Multicast IP.

To delete a paging group via phone user interface:

- 1. Press the paging list key when the phone is idle.
- **2.** Press \frown or \bigcirc to select a desired paging group.
- 3. Press the **Option** soft key, and then select **Delete** from the prompt list.

The LCD screen prompts "Delete selected paging group?".

	Paging List	
1. Product		
2. (Er 🚺 Warning		10,653,010
	te selected paging grou	up?
4. (Er		
5. (Empty)		
Cancel		ОК

4. Press the **OK** soft key to accept the change or the **Cancel** soft key to cancel.

If you want to delete all paging groups, you can press the **Delete All** soft key.

You can also configure the phone to use a default codec for sending multicast RTP stream via web user interface.

To configure a default codec for multicast paging via web user interface:

- 1. Click on Features->General Information.
- 2. Select the desired codec from the pull-down list of **Multicast Codec**.

The default codec is G722.

Yealink 1466							Eng	Log Out lish(English) -
	Status Account	Network	DSSKey	Featur	es	Settings	Directory	Security
Forward&DND	General Information	n					NOTE	
General Information	Call Waiting Call Waiting On Cod	P	Enabled	•	0		Call Waiting	ones to receive a
Audio	Call Waiting Off Cod				0			call when there is
Intercom	Auto Redial Auto Redial Interva	l (1~300s)	Disabled	-	0 0		Auto Redial It allows IP pho automatically re	
Transfer	Auto Redial Times (10		0		number after t	he first attempt.
Call Pickup	Key As Send		#	•	0		Assigns "#" or key.	"*" as the send
Remote Control Phone Lock			:				out the hotline	utomatically dial number when
ACD	DTMF Repetition		3	•	0		speakerphone key.	lset, pressing the key or the line
SMS	Multicast Codec		G722 Enabled	•	0		Call Completie It allows users	to monitor the
Action URL	Play Hold Tone Del	ау	30	•	0		busy party and when the busy available to rec	party becomes
Bluetooth	Hide Feature Acces		Disabled	•	0		You can cli	ck here to get
Power LED	Display Method on Auto Linekeys	Dialing	User Name	•	0		more guides.	5
Notification Popups	Confirm		Enabled	Cancel				

- 3. Click **Confirm** to accept the change.
- Note

If G722 codec is used for multicast paging, the LCD screen will display the icon 🚻 to indicate that it is providing high definition voice.

Default codec for multicast paging is configurable via web user interface only.

Receiving RTP Stream

You can configure the phone to receive a Real Time Transport Protocol (RTP) stream from the pre-configured multicast address(es) without involving SIP signaling. You can specify up to 31 multicast addresses that the phone listens to on the network.

Note

RTP stream is listened in the speakerphone mode by default. If you want to listen the RTP stream using the engaged audio device (speakerphone, handset or headset), contact your system administrator for more information.

Fixed volume to play RTP stream for specified paging group is configurable by your system administrator.

How the phone handles incoming multicast paging calls depends on Paging Barge and Paging Priority Active parameters configured via web user interface.

Paging Barge

The paging barge parameter defines the priority of the voice call in progress. If the priority of an incoming multicast paging call is lower than that of the active call, it will be ignored automatically. If Disabled is selected from the pull-down list of Paging Barge, the voice call in progress will take precedence over all incoming multicast paging calls. Valid values in the Paging Barge field:

- 1 to 31: Define the priority of the active call, 1 with the highest priority, 31 with the lowest.
- Disabled: The voice call in progress will take precedence over all incoming paging calls.

Paging Priority Active

The paging priority active parameter decides how the phone handles incoming multicast paging calls when there is already a multicast paging call on the phone. If enabled, the phone will ignore incoming multicast paging calls with lower priorities, otherwise, the phone will answer incoming multicast paging calls automatically and place the previous multicast paging call on hold. If disabled, the phone will automatically ignore all incoming multicast paging calls.

To configure multicast listening addresses via web user interface:

- 1. Click on Directory->Multicast IP.
- 2. Select the desired value from the pull-down list of Paging Barge.
- 3. Select the desired value from the pull-down list of Paging Priority Active.
- **4.** Enter the multicast IP address(es) and port number (e.g., 224.5.6.20:10008) which the phone listens to for incoming RTP multicast in the **Listening Address** field.
- 5. (Optional.) Enter the label in the Label field.

Label will appear on the LCD screen when receiving the multicast RTP stream.

6. Select the desired channel to listen from the pull-down list of Channel.

alink 1466	Status	Acco	unt Network	DSSKey	Features	Setting	,	Directory Security
Local Directory	Multicast List	ening						NOTE
Remote Phone		Paging I	Barge	10	• 🕜			Multicast Paging
Book		Paging I	Priority Active	Enabled	• 0			Multicast paging allows IP phone to send/receive Real-time
Phone Call Info	IP Addr	ess	Listening Address	Label	Chanr	el Priority		Transport Protocol (RTP) stream to/from the pre-configured
LDAP	1 IP Add	lress	224.5.6.20:10008	Product	0	• 1	ĥ	multicast address(es) without involving SIP signaling. Up to 10 listening multicast addresses car
Multicast IP	2 IP Add	lress			0	• 2	=	be specified on the IP phone.
Setting	3 IP Add	lress			0	• 3		You can click here to get
Setting	4 IP Add	lress			0	• 4		more guides.
	5 IP Add	lress			0	• 5		
	6 IP Add	lress			0	• 6		
	7 IP Add	lress			0	• 7		
	8 IP Add	lress			0	• 8		
	9 IP Add	lress			0	• 9		
	10 IP Ad	dress	_		0	• 10		

The default channel is 0.

7. Click **Confirm** to accept the change.

NoteThe priorities of listening addresses are predefined: 1 with the highest priority, 31 with the lowest.Multicast listening addresses are configurable via web user interface only.

Using Multicast Paging

To send RTP stream via a multicast paging key when the receiver's phone is idle:

1. Press the multicast paging key when the phone is idle.

The phone sends RTP to a preconfigured multicast address (IP: Port).

Both the sender's and receiver's phones play a warning tone and the receiver automatically answers the multicast RTP session in the speakerphone mode.

The multicast paging key LED illuminates solid green.

The following figure shows a multicast RTP session on the phone:



- 2. You can do the following:
 - To place the current multicast RTP session on hold, press the **Hold** soft key.

The sender's phone places the multicast RTP session on hold and receiver's phone releases the session.

To resume the held multicast RTP session, press the **Resume** soft key.

The multicast RTP session is established again.

• To end the multicast RTP session, press the **End Call** soft key.

Note Multicast RTP is one way only from the sender to the multicast address(es) (receiver). For outgoing RTP multicasts, all other existing calls on the phone will be placed on hold.

To send RTP stream via a paging list key when the receiver's phone is idle:

- **1.** Press the paging list key when the phone is idle.
- **2.** Press (\bullet) or (\bullet) to select the desired paging group.
- **3.** Press $(\circ \mathbf{K})$ or the **Paging** soft key to send RTP.
- 4. Your can do the following:
 - To place the current multicast RTP session on hold, press the Hold soft key.

The sender's phone places the multicast RTP session on hold and receiver's phone releases the session.

To resume the held multicast RTP session, press the **Resume** soft key.

The multicast RTP session is established again.

• To end the multicast RTP session, press the End Call soft key.

Music on Hold (MoH)

Music on hold (MoH) is the business practice of playing recorded music to fill the silence that would be heard by the party placed on hold. To use this feature, you should specify a SIP URI pointing to a Music on Hold Server account. When a call is placed on hold, the phone will send a SIP INVITE message to the Music on Hold Server account. The Music on Hold Server account automatically answers the SIP INVITE messages and immediately plays audio from some source located anywhere (LAN, Internet) to the held party. Contact your system administrator for the SIP URI.

Note If your server supports the MOH feature, you can also upload the custom music file into it directly. For more information, contact your service provider.

To configure music on hold server via web user interface:

- 1. Click on Account->Advanced.
- 2. Select the desired account from the pull-down list of Account.

3. Enter the SIP URI (e.g., sip:moh@sip.com) in the Music Server URI field.

Yealink 1466	Status Account Network	DSSKey Featu	res Settings	Log Out English(English) • Directory Security
Register	Account	Account 1		NOTE
	Keep Alive Type	Default 👻	0	
Basic	Keep Alive Interval(Seconds)	30	0	DTMF It is the signal sent from the IP
Codec	RPort	Disabled 🗸 🖓		phone to the network, which is generated when pressing the IP
Advanced	Subscribe Period(Seconds)	1800	0	phone's keypad during a call.
		÷		Session Timer It allows a periodic refresh of SIP sessions through a re-INVITE request, to determine whether a SIP
	SIP Server Type	Default 👻	0	session is still active.
	Music Server URI	sip:moh@sip.com	0	Busy Lamp Field/BLF List
	Directed Call Pickup Code	*97	0	Monitors a specific extension/a list of extensions for status
	Group Call Pickup Code	*98	0	changes on IP phones.
	Distinctive Ring Tones	Enabled 👻	0	Shared Call Appearance
	Unregister When Reboot	Disabled 👻	0	(SCA)/ Bridge Line Appearance (BLA)
	Out Dialog BLF	Disabled 👻	0	It allows users to share a SIP line on several IP phones. Any
	VQ RTCP-XR Collector name		0	IP phone can be used to originate or receive calls on the
	VQ RTCP-XR Collector address		0	shared line.
	VQ RTCP-XR Collector port	5060	0	Network Conference It allows multiple participants (more than three) to join in a

4. Click **Confirm** to accept the change.

When you place a call on hold, the held party can hear the music.

Note

For this feature to function, all involved parties cannot use encrypted RTP (SRTP). Music on hold server is configurable via web user interface only.

Automatic Call Distribution (ACD)

ACD is often used in offices for customer service, such as call center. The ACD system handles large volumes of incoming calls from callers who have no need to talk to a specific person but who require assistance from any of multiple persons at the earliest opportunity. ACD on the SIP-T46G IP phone allows the ACD system to distribute large volumes of incoming calls to the registered ACD users. To use this feature, you should configure an ACD key in advance. For more information on ACD key icon indicators, refer to Icon Instructions on page 3.

Note Make sure ACD is enabled on your IP phone. For more information, contact your system administrator.

To configure an ACD key via phone user interface:

- 1. Press Menu->Features->DSS Keys.
- 2. Select the desired DSS key.
- **3.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select **ACD** from the **Type** field.

4. (Optional.) Enter the string that will appear on the LCD screen in the Label field.



5. Press the Save soft key to accept the change or the Back soft key to cancel.

ACD key is configurable via web user interface at the path Dsskey->Line Key.

To log into the ACD system:

1. Press the ACD key when the phone is idle.

The LCD screen prompts you the following information:

User ID: the identity used to log into the queue.

Password: the password used to log into the queue.

	ACD	Login	
1. User ID:			
2. Password:			
Cancel	123	Delete	Login

- 2. Press the Login soft key to log in.
- Note Contact your system administrator for the User ID and Password to access the ACD system.

After configuring an ACD key, you can press the ACD key to log into the ACD system. After logging in, you are ready to receive calls from the ACD system. You can press the ACD key to show your current ACD user status. You can press the **Available/Unavailable** soft key to change your ACD user status. ACD user status synchronizes on both IP phone and ACD system.

When you set the ACD user status to be available, the ACD key LED illuminates solid green, and then the server begins distributing calls to your IP phone.

When you set the ACD user status to be unavailable, the ACD key LED flashes green, and then the server temporarily stops distributing calls to your IP phone. To log out of the ACD system, press the **Logout** soft key.

Note

It is recommended you configure no more than one ACD key per phone. At any time, at most one ACD key can be in progress on your phone.

Shared Call Appearance (SCA)

You can use SCA feature to share an extension which can be registered on two or more IP phones at the same time. The shared line is indicated by a different line icon.

4603
 13:39 Fri, Sep 02
 4603
 4604
 History Directory DND Menu

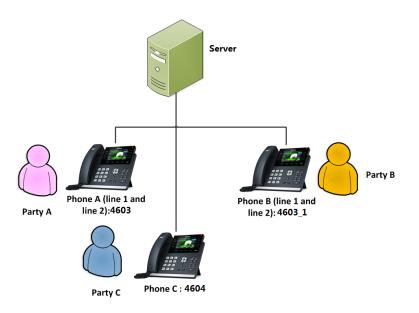
In the following figure, the first line is private and the second line is shared:

If two phones share a line, an incoming call to this extension will cause both phones to ring simultaneously. The incoming call can be answered on either phone but not both.

This feature is very useful in the boss and secretary scenario. For example, the secretary can share the boss's extension on her phone. When there is an incoming call to the extension of the boss, both the phones of the boss and the secretary will ring simultaneously. Either the boss or the secretary can answer the call. Calls on the shared line can be placed on hold, barged in or retrieved from another shared phone.

Configuring SCA Feature on the IP Phone

You can configure a primary account on the IP phone and other alternate accounts on the other IP phones. In a SCA Hybrid Key System scenario, you can automatically assign multiple DSS keys with Line type for a registered shared line on the phone (If auto linekeys feature is enabled). For example, party A, party B share the account 4603, phone A registers the primary account 4603 and assigns line key 1 and line key 2 for the account 4603, phone B registers the alternate account 4603_1 and assigns line key 1 and line key 2 for the account 4603_1, phone C registers the account 4604.



To configure the auto linekeys feature via web user interface:

- 1. Click on Features->General Information.
- 2. Select Enabled from the pull-down list of Auto LineKeys.

If **Auto LineKeys** is enabled, you can automatically assign multiple DSS keys with Line type for a registered shared line on the phone.

ealink 1466	Chattar	Account	Network	DSSKev	Featu	·06	Settings	Directory	Securit
	Status	Account	Network	DSSKey	reatu	C5	Settings	Directory	Security
Forward&DND	G	General Informati	on					NOTE	
General		Call Waiting		Enabled	•	0		Call Waiting	
Information		Call Waiting On Co	ode			0		It allows IP ph new incoming	
Audio		Call Waiting Off C	ode			0		already an act	
		Auto Redial		Disabled	•	0		Auto Redial	
Intercom		Auto Redial Inten	/al (1~300s)	10		0		It allows IP ph automatically r	edial a busy
Transfer		Auto Redial Times	s (1~300)	10		0		number after	
Call Pickup	Key As Send		#	•	0		Key As Send Assigns "#" or "*" as the send key.		
Remote Control								Hotline	
Phone Lock				:				IP phone will a out the hotlin	utomatically dia e number when
ACD		DTMF Repetition		3	•	0			dset, pressing t key or the line
SMS		Multicast Codec		G722	•	0		Call Complet	ion
		Play Hold Tone		Enabled	•	0		It allows users	to monitor the d establish a cal
Action URL		Play Hold Tone D	elay	30		0			y party become
Bluetooth		Hide Feature Acc	ess Codes	Disabled	•	0			
Power LED		Display Method or	n Dialing	User Name	•	0		You can cl more guides.	ick here to get
Notification Popups		Auto Linekeys		Enabled	•	0			

3. Click **Confirm** to accept the change.

To configure the shared line settings and the number of line key on phone A via web user interface:

1. Register the primary account 4603.

alink 146G	Status Account Networ		Features Settings	Lo English(English) Directory Securi
	Account	Account 1	•	NOTE
Register	Register Status	Registered	_	NOTE
Basic	Line Active	Enabled	•	Account Registration Registers account(s) for the
odec	Label	460.3		phone.
	Display Name	4603		Server Redundancy
Advanced			_	It is often required in VoIP deployments to ensure
	Register Name	4603		continuity of phone service, events where the server ne
	User Name	4603		to be taken offline for maintenance, the server fai
	Password	•••••		the connection between the
	SIP Server 1			phone and the server fails.
	Server Host	pbx.yealink.com	Port 5060	A general term for technique
	Transport	UDP	•	that establish and maintain connections traversing NAT
	Server Expires	3600		gateways. STUN is one of the NAT traversal techniques.
	Server Retry Counts	3		Hist dave sur centriques.
	SIP Server 2			You can configure NAT trave for this account.
	Server Host		Port 5060	for this account.
	Transport	UDP	•	You can click here to more guides.
	Server Expires	3600		more galacs.
	Server Retry Counts	3		
	·			
	Enable Outbound Proxy Server	Enabled	-	
	Outbound Proxy Server 1	10.1.8.11	Port 5060	
	Outbound Proxy Server 2		Port 5060	
	Proxy Fallback Interval	3600		
	NAT	Disabled	•	

- Click on Advanced, and then select Shared Call Appearance from the pull-down list of Shared Line.
- 3. Enter the desired number in the Number of line key field.

This field appears only if **Auto Linekeys** is enabled.

The default value is 1. In this example, the value is set to 2.

Yealink 1466				Log Out English(English) 🗸
	Status Account Netwo	ork DSSKey Featur	es Settings	Directory Security
Register	Account	Account 1 🗸	0	NOTE
Basic	Account Keep Alive Type Keep Alive Interval(Seconds) RPort	Default -	0	DTMF It is the signal sent from the IP
Codec		Disabled -	0	phone to the network, which is generated when pressing the IP phone's keypad during a call.
Advanced		:		phone's keypau duning a cail.
	BLF List Retrieve Call Parked Code	•	0	Session Timer It allows a periodic refresh of SIP sessions through a re-INVITE request, to
	BLF List Retrieve Call Parked Code Image: Construction of the construction o		session is still active.	
		*11 Disabled	0	Busy Lamp Field/BLF List Monitors a specific extension/a
	Distinctive Ring Tones	Enabled 👻	0	list of extensions for status changes on IP phones.
	Unregister When Reboot	Disabled 💌	0	Shared Call Appearance
	Out Dialog BLF	Disabled 👻	0	(SCA)/ Bridge Line Appearance (BLA)
	VQ RTCP-XR Collector name		0	It allows users to share a SIP line on several IP phones. Any
	VQ RTCP-XR Collector address		0	IP phone can be used to originate or receive calls on the
		5060	0	shared line.
	Number of line key	2 Cancel	0	Network Conference It allows multiple participants (more than three) to join in a call.

4. Click **Confirm** to accept the change.

The phone A will automatically configure the line keys from the first unused DSS key, according to the configured number in the **Number of line key** field.

To configure the shared line settings and the number of line key on phone B via web user interface:

1. Register the alternate account 4603_1.

alink 146G	Status	Account	Network	DSSKey	Features	Settings	Directory	Security	
Register	Acc	ount		Account 1	•		NOTE		
	Reg	ister Status		Registered					
Basic	Line	Active		Enabled	•			unt(s) for the IF	
Codec	Labe	el		4603_1			phone.		
Advanced	Disp	lay Name		4603_1			Server Redu It is often requ		
	Reg	ister Name		4603			deployments to continuity of pl	o ensure hone service, fo	
	User	r Name		4603_1			events where to be taken off	the server need fline for	
	Pass	sword		•••••				the server fails, between the IF	
	SIP	Server 1					phone and the		
	Serv	ver Host		pbx.yealink.com	Port	5060	NAT Travers	al for techniques	
	Tran	nsport		UDP	•			and maintain IP	
	Serv	Server Expires					gateways. STUN is one of the NAT traversal techniques. You can configure NAT travers		
	Serv	er Retry Counts		3					
	SIP	SIP Server 2							
	Serv	ver Host			Port	5060	for this account.		
	Tran	nsport		UDP	•		You can c more guides.	lick here to ge	
	Serv	/er Expires		3600					
	Serv	ver Retry Counts		3					
	Enal	ble Outbound Proxy	Server	Enabled	•				
	Out	bound Proxy Server	1	10.1.8.11	Port	5060			
	Out	bound Proxy Server	2		Port	5060			
	Prox	ky Fallback Interval		3600					
	NAT			Disabled					

(Enter the primary account 4603 in the Register Name field.)

- 2. Click on Advanced, select Shared Call Appearance from the pull-down list of Shared Line.
- 3. Enter the desired number in the **Number of line key** field.

This field appears only if **Auto Linekeys** feature is enabled.

The default value is 1. In this example, the value is set to 2.

ealink 1466	Status Account Network	DSSKey Featur	res Settings	Directory	
Register	Account	Account 1	Account 1		
Basic	Keep Alive Type	Default 👻	0	D.T.M.F.	
Dasic	Keep Alive Interval(Seconds)	30	0	DTMF It is the signal sent from t	
Codec	RPort	Disabled 👻	0	phone to the network, w generated when pressing	
Advanced				phone's keypad during a c	
		:		Session Timer It allows a periodic refresh	
	BLF List Retrieve Call Parked Code		0	SIP sessions through a	
	Shared Line	Shared Call Appearance 👻	0	re-INVITE request, to determine whether a SIP session is still active.	
	Call Pull Feature Access Code	*11	0	Session is sell deeve.	
	Dialog Info Call Pickup	Disabled -	0	Busy Lamp Field/BLF List Monitors a specific extensio list of extensions for status changes on IP phones.	
	Distinctive Ring Tones	Enabled -	0		
	Unregister When Reboot	Disabled -	õ		
	Out Dialog BLF	Disabled -	õ	Shared Call Appearance (SCA)/ Bridge Line	
	VO RTCP-XR Collector name		ñ	Appearance (BLA) It allows users to share a s	
	VQ RTCP-XR Collector address		õ	line on several IP phones. IP phone can be used to	
	VQ RTCP-XR Collector port	5060	õ	originate or receive calls or shared line.	
	Number of line key	2	0	Network Conference	

4. Click **Confirm** to accept the change.

The phone B will automatically configure the line keys from the first unused DSS key, according to the configured number in the **Number of line key** field.

Configuring private hold soft key or private hold key

Public hold allows any shared line to retrieve the held call. Private hold only allows the hold party to retrieve the held call. For example, you can retrieve the held call on either phone A or phone B when you place a call on public hold; you can retrieve the held call only on phone A when you place a call on private hold on phone A. By default, the private hold soft key isn't displayed on the LCD screen. You need to configure either the private hold soft key or a private hold key before you place the call on private hold.

To configure the private hold soft key via web user interface:

- 1. Click on Settings->Softkey Layout.
- 2. Select Enabled from the pull-down list of Custom Softkey.
- 3. Select On Talk from the pull-down list of Call States.
- Select Private Hold from the Unselected Softkeys column and then click → .
 The Private Hold appears in the Selected Softkeys column.

Yealink 1466	Log Out English(English) V				
	Status Account Network DSSKey Features Settings	Directory Security			
Preference Time & Date	Custom Softkey Enabled	NOTE Softkey Layout			
Call Display	Call States On Talk Q	It is used to customize the soft keys at the bottom of the LCD screen to best meet users' requirements.			
Upgrade Auto Provision	Empty Transfer Mute Hold Swap Conference	You configure the softkey layout using the softkey layout templates for different call states.			
Configuration Dial Plan	New Call → End Call ↑ Switch Account Answer Reject ← ↓	You can click here to get more guides.			
Voice					
Ring Tones	Confirm Cancel Reset to default				
Softkey Layout					

5. Click Confirm to accept the change.

Note Configuring the private hold soft key may affect the softkey layout in the Talking state. Contact your administrator for more information.

To configure a private hold key via phone user interface:

- 1. Press Menu->Features->DSS Keys.
- 2. Select the desired DSS key.
- **3.** Press (\cdot) or (\cdot) , or the **Switch** soft key to select **Key Event** from the **Type** field.

4. Press (•) or (•), or the **Switch** soft key to select **Private Hold** from the **Key Type** field.

Line Key 1						
1. Type:	Key Event	$\triangleleft \triangleright$				
2. Key Type:	Private Hold	$\triangleleft \triangleright$				
3. Label:						
Back	Switch	Save				

- 5. (Optional.) Enter the string that will appear on the LCD screen in the Label field.
- 6. Press the Save soft key to accept the change or the Back soft key to cancel.

Configuring call pull feature

Call pull feature allows users to retrieve an existing call from another shared phone that is in active or public hold status.

To configure the call pull feature access code via web user interface:

- 1. Click on Account->Advanced.
- 2. Select the desired account from the pull-down list of Account.
- 3. Enter the call pull feature access code (e.g., *11) in the Call Pull Feature Access Code field.

Yealink 1466						Log Out ish(English) 🗸		
	Status Account	Network	DSSKey	Features	Settings	Directory	Security	
Register	Account		Account 1 🔹 ?			NOTE		
Basic	Keep Alve Type Default ? Keep Alve Interval(Seconds) 30 ? RPort Disabled ? BLF List Retrieve Call Parked Code ? ? Shared Line Shared Call Appearance ? Call Pull Feature Access Code *11 ? Dialog Info Call Pickup Disabled ? Distinctive Ring Tones Enabled ? Unregister When Reboot Disabled ? Out Dialog BLF Disabled ? VQ RTCP-XR Collector name ? ? VQ RTCP-XR Collector port 5060 ?					DTMF It is the signal sent from the IP phone to the network, which is		
Codec			generated whe	generated when pressing the IP phone's keypad during a call.				
				Shared Call Appearance 🗸 🥜		Session Timer It alows a periodic refresh of SIP sessions through a re-INVITE request, to determine whether a SIP session is still active.		
			Disabled	• ?		Busy Lamp Field/BLF List Monitors a specific extension/a list of extensions for status changes on IP phones.		
				• 0		Shared Call Appearance (SCA)/ Bridge Line Appearance (BLA) It allows users to share a SIP line on several IP phones. Any IP phone can be used to orginate or receive calls on the shared line.		
			5060					
	Number of line key		2	Cancel		Network Conference It allows multiple participants (more than three) to join in a call.		

4. Click **Confirm** to accept the change.

The phone will dial out "*11" automatically when you press the Call Pull soft key.

Using SCA Feature on the IP Phone

This section provides you with detailed information on using the SIP-T46G IP phone in a SCA Hybrid Key System scenario. In a SCA Hybrid Key System scenario, the status of the line key LED which associates with a shared line will change. For more information on line key LED indicators, refer to LED Instructions on page 9.

You can do the following using the IP phone in a SCA Hybrid Key System scenario:

- Placing calls
- Answering calls
- Placing a call on hold
- Retrieving a held call
- Barging in an active call
- Call Pull

Placing Calls

You can have one call or multiple calls on the shared line.

To place a call on the shared line:

Do one of following:

Enter the desired number using the keypad when the phone is idle. Press (or), (# ===), or the Send soft key.

The phone will dial the entered number using the first line key.

- Press the line key when the phone is idle to enter the dialing screen.

Enter the desired number using the keypad and then press $(\mathbf{v}, | \mathbf{f}_{\mathbf{k},\mathbf{v}} |$, or the **Send** soft key.

The phone will dial the entered number using the selected line key.

To place multiple calls on the shared line:

You can have more than one call on the shared line. To place a new call when there is an active call on the line key 1 of phone A, do one of the following on phone A:

- Press 🐻 or the **Hold** soft key. The original call is placed on hold.

Press the New Call soft key to enter the dialing screen.

Enter the desired number using the keypad.

Press	(ок`),	# SEND	or	the	Send	soft	key.
-------	-------	----	--------	----	-----	------	------	------

Phone A will dial the entered number using the line key 2 automatically.

- Press the line key. The original call is placed on hold.

Enter the desired number using the keypad.

Press (or , #send), or the **Send** soft key.

Phone A will dial the entered number using the selected line key.

Answering Calls

You can have one call or multiple calls on the shared line. Incoming calls will be distributed evenly among the available line keys.

To answer a call on the shared line:

When an incoming call arrives on the shared line, the phone A and phone B will ring simultaneously, and the LED indicators of the line key 1 on both phone A and phone B will flash green. You can answer the incoming call on either phone A or phone B but not both.

Do one of the following on phone A or phone B:

- Press the line key 1, $(\bullet \kappa)$, $(\circ \kappa)$ or the **Answer** soft key on phone A.

The LED indicator of the line key 1 on phone A will illuminate solid green and phone B will illuminate solid Red.

- Press the line key 1, $\boxed{\quad \blacksquare}$, $(\ensuremath{\circ}\ensuremath{\circ}\ensuremath{\circ}$) or the **Answer** soft key on phone B.

The LED indicators of the line key 1 on phone A will illuminate solid red and phone B will illuminate solid green.

To answer multiple calls on the shared line:

An incoming call arrives on the shared line when there is an active call on phone A's line key 1. The LED indicators of the line key 2 on both phone A and phone B will fast flash green. You can answer the incoming call on either phone A or phone B. The LCD screen of phone A displays the information of the incoming call (e.g., "Incoming Call: pbx.yealink.com").



Note Make sure call waiting feature is enabled on phone A. For more information, refer to Call Waiting on page 146.

Do one of the following on phone A:

- Press the line key 2. Phone B stops ringing.
- Press the Answer soft key. Phone B stops ringing.
- Press (\bullet) to access the new call.

Press (∞) or the **Answer** soft key. Phone B stops ringing.

The incoming call is answered and the original call is placed on hold. The LED indicators of the line key 1 on phone A will slowly flash green and phone B will slowly flash red indicating that there is the held call on the line key 1. The LED indicators of the line key 2 on phone A will illuminate solid green and phone B will illuminate solid red indicating that there is an active call on the line key 2.

You can also answer the call on phone B. Do one of the following on phone B:

- Press the line key 2. Phone A stops ringing.
- Press (or) or the **Answer** soft key. Phone A stops ringing.

The LED indicators of the line key 2 on phone A will illuminate solid red and phone B will illuminate solid green indicating that there is an active call on the line key 2. Meanwhile, The LED indicators of the line key 1 phone A will illuminate solid green and phone B will illuminate solid red indicating that there is an active call on the line key 1.

Note If the number of incoming calls is greater than the configured line keys, the line keys will be used by sequence circulation.

Placing a Call on Hold

To place a call on public hold:

1. Press **[d**] or the **Hold** soft key on phone A when party A and party C are talking.



The line key LEDs on phone A slowly flash green and phone B slowly flash red when the shared line call is placed on hold.

To place a call on private hold:

 Press the **PriHold** soft key or private hold key on phone A when there is an active call on the shared line (You may need to press the **More** soft key to see the **PriHold** soft key).



The line key LED flashes green slowly on phone A when the shared line call is placed on private hold.

The line key LED illuminates solid red on phone B when the shared line call is placed on private hold.

Retrieving a Held Call

To retrieve a call placed on public hold:

You can retrieve the public held call on either phone A or phone B.

To retrieve the call on phone A:

1. Press the line key with the slowly flashing green LED indicator, or the **Resume** soft key.

The conversation between phone A and phone C is retrieved.

To retrieve the call on phone B:

Do one of the following:

- Press the line key with the slowly flashing red LED indicator.
- Long press the desired line key.
 - The Cancel, Call Pull, New Call and Retrieve soft keys appear on the LCD screen.



Press the Retrieve soft key to retrieve the call.

The conversation is established between phone B and phone C, phone A disconnects the call. And the line key LEDs on phone A illuminate solid red and phone B illuminate solid green.

To retrieve a call placed on private hold:

The private held call can be only retrieved by the hold party (party A).

Do one of the following:

- Press the line key with the slowly flashing green LED indicator.
- Press the **Resume** soft key on phone A.

Barging In an Active Call

To interrupt the active call on the shared line:

If phone A has only one active call, do the following:

1. Long press the desired line key on phone B.

The **Cancel**, **Call Pull**, **New Call** and **BargeIn** soft keys appear on the LCD screen of phone B.

	Share	ed Calls	1/1	
1. SCA Active : 4604 Yealink				
Cancel	Call Pull	New Call	Retrieve	

2. Press the BargeIn soft key to interrupt the active call of phone A.

Party B will set up a conference call with the other parties in the active call.

- **3.** Press **(**, the **Hold** or the **End Call** soft key.
 - If any party in the conference call presses or the **Hold** soft key, two-way voice can be heard between the remaining parties.
 - If party A or party B presses the **End Call** soft key, the remaining parties are still connected. If the other party (not the shared line party) presses the **End Call** soft key, the conference call is ended.

You can also press the line key with the solid red LED indicator on phone B to set up a conference call with the other parties in the active call.

If phone A has more than one call, do the following:

1. Long press the desired line key on phone B.

The list of calls appears on the LCD screen of phone B.

	Shared Calls			
1. SCA Held : 4604 Yealink				
2. SCA Active	e : 4609 Yealink	(
Coursel	C-11 D-11	No.	Dations	
Cancel	Call Pull	New Call	Retrieve	

2. Press () or () to select the active call.

The **Cancel**, **Call Pull**, **New Call** and **BargeIn** soft keys appear when the active call is highlighted.

	2/2			
1. SCA Held : 4604 Yealink				
2. SCA Active : 4609 Yealink				
Cancel	Call Pull	New Call	Barge In	

3. Press the BargeIn soft key to interrupt the active call of phone A.

Party B may hear a warning tone and then set up a conference call with the other parties of the active call.

- 4. Press , the Hold or the End Call soft key.
 - If any party in the conference call presses .
 be heard between the remaining parties.
 - If party A or party B presses the **End Call** soft key, the remaining parties are still connected. If the other party (not the shared line party) presses the **End Call** soft key, the conference call is ended.

Call Pull

Call pull feature allows users to retrieve an existing call from another shared phone that is in active or hold status. For example, when there is a call between phone A and phone C, you can use call pull feature on phone B to retrieve this call from phone A. Then the call is established between phone B and phone C.

To retrieve a call from another shared phone:

If there is an active call between phone A and phone C, do the following:

- 1. Long press the desired line key on phone B.
 - The **Cancel**, **Call Pull**, **New Call** and **BargeIn** soft keys appear on the LCD screen of phone B.



2. Press the Call Pull soft key.

The active call has been retrieved from the phone A successfully.

If there is a held call between phone A and phone C, do the following:

1. Long press the desired line key on phone B.

The **Cancel**, **Call Pull**, **New Call** and **Retrieve** soft keys appear on the LCD screen of phone B.



2. Press the Call Pull soft key.

The held call has been retrieved from the phone A successfully.

Bridged Lines Appearance (BLA)

BLA allows users to share a SIP line on two or more IP phones. Users can monitor the specific extension (BLA number) for status changes on each IP phone. To use this feature, a BLA group should be pre-configured on the server and one of them is specified as a BLA number. BLA depends on support from a SIP server.

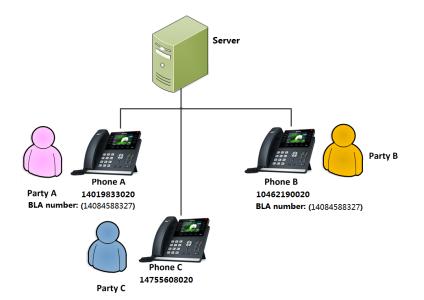
140845883	27	13	:39 Fri, Sep 02
<u>6</u> 140845			
104621			
History	Directory	DND	Menu

In the following figure, the first line is private and the second line is shared:

Any IP phone can be used to originate or receive calls on the bridged line. An incoming call to the BLA number can be presented to multiple phones in the group simultaneously. The incoming call can be answered on any IP phone of the group but not all.

Configuring BLA Feature on the IP Phone

You can share a BLA number on two or more phones. For example, phone A registers the account 14019833020 and assigns BLA number, phone B registers the account 10462190020 and assigns BLA number, phone C registers the account 14755608020. Phone A and phone B share the BLA number 14084588327.



To register an account and configure BLA feature on phone A via web user interface:

1. Register the account 14019833020.

alink 146G						Eng	glish(English)
	Status Accour	it Networ	k DSSKey F	eatures	Settings	Directory	Security
Register	Account		Account 1	- 7		NOTE	
De ete	Register Status		Registered			Account Reg	
Basic	Line Active		Enabled	- 0		Registers acco	
Codec	Label		14019833020	0		phone.	
Advanced	Display Name		14019833020	0		Server Redu It is often requ	ired in VoIP
	Register Name		14019833020	0		deployments to continuity of pl	
	User Name		14084588327	0		events where t to be taken off	
	Password		•••••	0		maintenance, t the connection	he server fails, between the If
	SIP Server 1	0				phone and the	
	Server Host		sip.ringcentral.com	Port 5060	. O	NAT Travers	al for techniques
	Transport		UDP	v 🕜		that establish and maintain connections traversing NAT	and maintain IP
	Server Expires		3600	0		gateways. STL NAT traversal	IN is one of the
	Server Retry Cou	nts	3	0		NAT traversar	ecriniques.
	SIP Server 2	0				You can config for this accoun	
	Server Host			Port 506	0 0	for this account	t.
	Transport		UDP	- 0		You can c more guides.	lick here to g
	Server Expires		3600	0			
	Server Retry Cou	Server Retry Counts		0			
	Enable Outbound	Proxy Server	Enabled	• Ø			
	Outbound Proxy S	Gerver 1	sip114.ringcentral.com	Port 509	9 🕜		
	Outbound Proxy S	Server 2		Port 5060) ()		
	Proxy Fallback Int	erval	3600	0			
			Disabled	-			

- 2. Click on Advanced, and then select Draft BLA from the pull-down list of Shared Line.
- 3. Enter the desired number in the **BLA Number** field.

ealink 1466					Eng	Log (ish(English)
	Status Account Network	DSSKey	eatures	Settings	Directory	Security
Register	Account	Account 1	• ?		NOTE	
Basic	Keep Alive Type	Default	• 🕜			
Basic	Keep Alive Interval(Seconds)	30	0		DTMF It is the signal s	
Codec	RPort	Disabled	- 0		phone to the r generated whe	
Advanced		_			phone's keypad	during a call.
	BI F List Retrieve Call Parked Code		0		Session Timer It allows a periodic refresh o SIP sessions through a re-INVITE request, to determine whether a SIP session is still active.	
	Shared Line	Draft BLA	- 0			
	Dialog Info Call Pickup	Disabled	• 🕜		Busy Lamp Fie	
	BLA Number	14084588327	0		Monitors a spec list of extension	
	BLA Subscription Period	300	0		changes on IP	phones.
	SIP Send MAC	Disabled	• 🕜		Shared Call A	000000000
	Out Dialog BLF	Disabled	- 0		(SCA)/ Bridge	Line
	VQ RTCP-XR Collector name		0		Appearance (It allows users	to share a SIP
	VQ RTCP-XR Collector address		0		line on several IP phone can b	e used to
	VO RTCP-XR Collector port	5060	0		originate or rec shared line.	eive calls on th
	Number of line key	2	0			
	Confirm	Cano			Network Cont It allows multip (more than thr call.	e participants

4. Click **Confirm** to accept the change.

To register an account and configure BLA feature on phone B via web user interface:

1. Register the account 10462190020.

· · · · ·			Log Out English(English) 🗸
Yealink 1466	Status Account Network	DSSKey Features Settings	Directory Security
_			Directory
Register	Account	Account 1 👻 🕜	NOTE
Basic	Register Status	Registered	Account Registration
bush	Line Active	Enabled 🗸 🥑	Registers account(s) for the IP phone.
Codec	Label	10462190020	
Advanced	Display Name	10462190020	Server Redundancy It is often required in VoIP
	Register Name	10462190020	deployments to ensure continuity of phone service, for
	User Name	14084588327 🕜	events where the server needs to be taken offline for
	Password	••••••	maintenance, the server fails, or the connection between the IP
	SIP Server 1 🕜		phone and the server fails.
	Server Host	sip.ringcentral.com Port 5060	NAT Traversal A general term for techniques
	Transport	UDP 🗸 💡	that establish and maintain IP connections traversing NAT
	Server Expires	3600	gateways. STUN is one of the NAT traversal techniques.
	Server Retry Counts	3	NAT daversar techniques.
	SIP Server 2 🕜		You can configure NAT traversal for this account.
	Server Host	Port 5060	
	Transport	UDP 👻 😮	You can click here to get more guides.
	Server Expires	3600	
	Server Retry Counts	3	
		Enabled 🗸 🖉	
	Enable Outbound Proxy Server		
	Outbound Proxy Server 1	sip214.ringcentral.com Port 5099	
	Outbound Proxy Server 2	Port 5060	
	Proxy Fallback Interval	3600	
	NAT	Disabled 🗸 🥥	
	Confirm	Cancel	

- 2. Click on Advanced, and then select Draft BLA from the pull-down list of Shared Line.
- 3. Enter the desired number in the **BLA Number** field.

Yealink T466				Log Out English(English) 🗸
	Status Account Network	DSSKey Fe	atures Settings	Directory Security
Register	Account	Account 1	• ?	NOTE
Basic	Keep Alive Type	Default	• ()	DTMF
Basic	Keep Alive Interval(Seconds)	30	0	It is the signal sent from the IP
Codec	RPort	Disabled	• 🕜	phone to the network, which is generated when pressing the IP
Advanced				phone's keypad during a call.
	BLF List Retrieve Call Parked Code	:	0	Session Timer It allows a periodic refresh of SIP sessions through a re-INVITE request, to determine whether a SIP
	Shared Line	Draft BLA	▼ ②	session is still active.
	Dialog Info Call Pickup	Disabled	▼ (2)	
	BLA Number	14084588327	0	Busy Lamp Field/BLF List Monitors a specific extension/a
	BLA Subscription Period	300	0	list of extensions for status changes on IP phones.
	SIP Send MAC	Disabled	• 🕜	
	Out Dialog BLF	Disabled	- 0	Shared Call Appearance (SCA)/ Bridge Line
	VO RTCP-XR Collector name			Appearance (BLA) It allows users to share a SIP
	VO RTCP-XR Collector address			line on several IP phones. Any IP phone can be used to
	VO RTCP-XR Collector port	5060	0	originate or receive calls on the shared line.
	Number of line key	2		silared mile.
	Confirm	2 Cance	el	Network Conference It allows multiple participants (more than three) to join in a call.

4. Click **Confirm** to accept the change.

Using BLA Feature on the IP Phone

This section provides you with detailed information on using the SIP-T46G IP phone in a BLA scenario. In a BLA scenario, the status of the line key LED which associates with a bridged line will change. For more information on line key LED indicators, refer to Icon Instructions on page 3.

You can do the following using the IP phone in a BLA scenario:

- Placing calls
- Answering calls
- Place a call on hold
- Retrieving a held call

Placing Calls

You can have one call or multiple calls on the bridged line.

To place a call on the bridged line:

Do one of following:

Enter the desired number using the keypad when the phone is idle. Press (or), # , or the Send soft key.

The phone will dial the entered number.

- Press the line key when the phone is idle to enter the dialing screen.

Enter the desired number using the keypad and then press (∞) , $(\#_{wo})$, or the **Send** soft key.

The phone will dial the entered number.

To place multiple calls on the bridged line:

You can have more than one call on the bridged line. To place a new call where there is an active call on phone A, do one of the following on the phone A:

Press 🕼 or the **Hold** soft key. The original call is placed on hold.

Press the New Call soft key to enter the dialing screen.

Enter the desired number using the keypad.

Press (o_{κ}) , $|| \#_{SEND}|$, or the **Send** soft key.

Phone A will dial the entered number.

- Press the line key. The original call is placed on hold.

Enter the desired number using the keypad.

Press $(\mathbf{o}\mathbf{K})$, $|| \#_{\text{SEND}}||$, or the **Send** soft key.

Phone A will dial the entered number.

Answering Calls

When the phone C dials the BLA number "14084588327", an incoming call will arrive on the bridged lines. The phone A and phone B ring simultaneously and the LED indicators of the line key on both phone A and phone B will flash green. You can answer the incoming call on either phone A or phone B but not both.

Do one of the following on phone A or phone B:

- Press (ок) , 💽 or the **Answer** soft key on phone A.

Phone B stops ringing. The LED indicator of the line key on the phone A will illuminate solid green. At the meanwhile, the LED indicator of the line key on the phone B will illuminate solid red indicating that there is an active call on the phone A.

- Press (ок), () or the **Answer** soft key on phone B.

Phone A stops ringing. The LED indicator of the line key on the phone B will illuminate solid green. At the meanwhile, the LED indicator of the line key on the phone A will illuminate solid red indicating that there is an active call on the phone B.

Placing a Call on Hold

To place a call on hold:

Press or the Hold soft key on phone A when party A and party C are talking.
 The line key LEDs on phone A slowly flash green and phone B slowly flash red when the bridged line call is placed on hold.

Retrieving a Held Call

If there is a held call between phone A and phone C, you can retrieve a held call on either phone A or phone B.

To retrieve the held call on phone A:

1. Press the line key with the slowly flashing green LED indicator, or the **Resume** soft key on the phone A.

The conversation between phone A and phone C is retrieved.

To retrieve the held call on phone B:

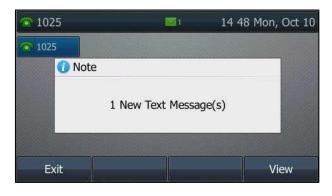
1. Press the line key with the slowly flashing red LED indicator on phone B.

The conversation is established between phone B and phone C, phone A disconnects the call.

Messages

Short Message Service (SMS)

You can send and receive text messages using the SIP-T46G IP phone. New text messages can be indicated both acoustically and visually. When receiving a new text message, the phone will play a warning tone. The power indicator LED will slowly flash red, and the LCD screen will prompt "n New Text Message(s)" ("n" indicates the number of unread text messages. e.g., 1 New Text Message(s)) and an icon



Note When the phone receives a text message, the text message prompt window will pop up by default, if you want to disable the feature, contact your system administrator for more information.

You can store text messages in your phone's Inbox, Sentbox, Outbox or Draftbox. Each of the boxes can store up to 100 text messages. If the number of the text messages in one box is more than 100, the phone will directly delete the oldest text message in the box.

Note SMS is not available on all servers. Contact your system administrator for more information.

To read a text message:

1. Press Menu->Message->Text Message->Inbox.

Int	xoo	Sentbox	Outbo	•X 1/4 ►
103		Today	14:21	
2 103		Today	14:20	
Back	Rep	oly O	ption	View

- 2. Select the desired message and then press the **View** soft key.
- Note If the phone prompts receiving new text messages, you can also press the View soft key to read the new messages directly.

To send a text message:

- 1. Press Menu->Message->Text Message->New Message.
- 2. Compose the new text message.

You can press the **abc** soft key to change the input mode.

	New M	lessage	
hi,			
Back	abc	Delete	Send

- 3. Press the Send soft key.
- **4.** (Optional.) Press (•) or (•), or the **Switch** soft key to select the desired account from the **From** field.
- 5. Enter the number you want to send the message to in the **To** field.
- 6. Press the Send soft key to send the message or the Back soft key to cancel.

Sending a text message is configurable via web user interface at the path Features->SMS.

To reply a text message:

- 1. Press Menu->Message->Text Message->Inbox.
- 2. Select the desired message and then press the **Reply** soft key.
- 3. Compose the new text message. You can press the **abc** soft key to change the input mode.

To:1026<1026@10.2.1.48>					
hi					
Back	abc	Delete	Send		

4. Press the Send soft key after completing the content.

5. Check the From and To fields, and then press the Send soft key.

To delete a text message:

- 1. Press Menu->Message->Text Message->Inbox (Sentbox, Outbox or Draftbox).
- 2. Select the desired message and then press the **Option** soft key.
- 3. Select Delete to delete the desired message, then press OK.

The LCD screen prompts "Delete the selected message?".

¥	Inbox	Sentbox	Outbox	1/4 🕨
	103	Today	14:21	
	1 🚺 Note			
	Dele	te the selected m	nessage?	
Cancel				ОК

4. Press the **OK** soft key to delete this message or the **Cancel** soft key to cancel.

You can also delete all text messages by pressing the **Option** soft key and then select **Delete All**. For more information, refer to the above steps.

Note You can also delete a specific message by pressing the Delete soft key after viewing.

Voice Mail

You can leave voice mails for someone else using the SIP-T46G IP phone. You can also listen to voice mails that are stored in a voice mailbox. This feature is set up on the server side. It is not available on all servers.

When receiving a new voice mail, the phone will play a warning tone. The power indicator LED will slowly flash red, and the LCD screen will prompt "n New Voice Mail(s)" ("n" indicates the number of unread voice messages, e.g., 2 New Voice Mail(s)) and an icon



If the voice mail pop-up message box disappears, it won't pop up again unless the user receives a new voice mail or the user re-registers the account that has unread voice mail(s).

Note You can configure the phone not to display the pop-up prompt, contact your system administrator for more information.

To leave a voice mail:

You can leave a voice mail for someone else when he/she is busy or inconvenient to answer the call. Follow the voice prompt from the system server to leave a voice mail, and then hang up.

To configure voice mail access codes via phone user interface:

- 1. Press Menu->Message->Voice Mail->Set Voice Mail Code.
- 2. Press the navigation keys to highlight the account which you want to set.
- **3.** Press the **123** soft key to select the proper input mode and then enter the voice mail access code (e.g., *4).

<u>.</u>		Mail Code	
1. 101:		*4	
2. 1007:			
Back	123	Delete	Save

4. Press the Save soft key to accept the change or the Back soft key to cancel.

Note Voice mail access codes must be predefined on the system server. Contact your system administrator for the more information.

To listen to voice mails:

- When the LCD screen prompts that the phone receives a new voice mail and the power indicator LED slowly flashes red, you can press or the **Connect** soft key to dial out the voice mail access code.
- 2. Follow the voice prompt to listen to your voice mails.

Note Before listening to voice mails, make sure the voice mail access code has been configured. When all new voice mails are retrieved, the power indicator LED will go out. To view the voice mail via phone user interface:

1. Press Menu->Message->Voice Mail->View Voice Mail.

The LCD screen displays the amount of new and old voice mails.

View Voice Mail			
1. 101 2 new 647 old mail			
2. 1007 0 new 0 old mail			
Back	Connect		

2. Select an account and then press the **Connect** soft key to listen to voice mails.

Message Waiting Indicator (MWI)

The SIP-T46G IP phone supports MWI when receiving a new voice message. If someone leaves you a voice mail, you will receive a message waiting indicator. MWI will be indicated in three ways: a warning tone, an indicator message (including a voice mail icon) on the LCD screen, and the power indicator LED slowly flashes red. This will be cleared when you retrieve all voice mails or delete them.

The MWI service is unsolicited for some servers, so the SIP-T46G IP phone only handles the MWI messages sent from the server. But for other servers, the MWI service is solicited, so the SIP-T46G IP phone must enable subscription for MWI.

Note MWI service is not available on all servers. Contact your system administrator for more information.

Options	Description		
Subscribe for MWI	Enable or disable a subscription for MWI service.		
MWI Subscription Period	Period of MWI subscription. The IP phone sends a refresh SUBSCRIBE request before initial SUBSCRIBE expiration.		
Subscribe MWI To Voice Mail	Enable or disable a subscription to the voice mail number for MWI service. To use this feature, you should also configure the voice mail number.		

The MWI subscription parameters you need to know:

Note

The phone will send SUBSCRIBE messages for the MWI service to the account or the voice number MWI service depending on the server. Contact your system administrator for more information.

To configure subscribe for MWI via web user interface:

- 1. Click on Account->Advanced.
- 2. Select the desired account from the pull-down list of Account.
- 3. Select Enabled from the pull-down list of Subscribe for MWI.
- 4. Enter the period time in the MWI Subscription Period(Seconds) field.

Vortintel			Log Out English(English) 🗸
Yealink 1466	Status Account Network	DSSKey Features	Settings Directory Security
Register	Account	Account 1 🗸 🦿	NOTE
Basic	Keep Alive Type	Default 🗸 💙	DTMF
50510	Keep Alive Interval(Seconds)	30	It is the signal sent from the IP phone to the network, which is
Codec	RPort	Disabled 🗸 🥥	generated when pressing the IP phone's keypad during a call.
Advanced	Subscribe Period(Seconds)	1800	phone's keypad during a cail.
	DTMF Type	RFC2833 🗸 💡	Session Timer It allows a periodic refresh of SIP
	DTMF Info Type	DTMF-Relay 🗸 🥐	sessions through a re-INVITE request, to determine whether a
	DTMF Payload Type(96~127)	101 🥜	SIP session is still active.
	BLF Send DTMF	Disabled 🗸	Busy Lamp Field/BLF List
	BLF DTMF Code		Monitors a specific extension/a list of extensions for status
	Retransmission	Disabled 🗸 🥜	changes on IP phones.
	Subscribe Register	Disabled 🗸 🥐	Shared Call Appearance
	Subscribe for MWI	Enabled 🗸 🕜	(SCA)/ Bridge Line Appearance (BLA)
	MWI Subscription Period(Seconds)	3600	It allows users to share a SIP line on several IP phones. Any IP
	Subscribe MWI To Voice Mail	Disabled 🗸 🥥	phone can be used to originate or receive calls on the shared
	Voice Mail	0	line.
	Voice Mail Display	Enabled V	Network Conference

5. Click **Confirm** to accept the change.

The IP phone will subscribe to the account number for MWI service by default.

To enable subscribe MWI to voice mail via web user interface:

- 1. Click on Account->Advanced.
- 2. Select the desired account from the pull-down list of Account.
- 3. Select Enabled from the pull-down list of Subscribe for MWI.
- 4. Select Enabled from the pull-down list of Subscribe MWI To Voice Mail.

5. Enter the desired voice mail number in the Voice Mail field.

ealink 146g		English(English				
Ealink 146G	Status Account Netwo	ork DSSKey	Features	Settings	Directory	Security
Register	Account	Account 1	✓ ?		NOTE	
De sta	Keep Alive Type	Default	∨ ?		DTHE	
Basic	Keep Alive Interval(Seconds)	30	0		DTMF It is the signal sent from the phone to the network, which generated when pressing the phone's keypad during a call.	
Codec	RPort	Disabled	✓ ()			
Advanced	Subscribe Period(Seconds)	1800	0			
	DTMF Type	RFC2833	✓ 🕜		Session Timer	
	DTMF Info Type	DTMF-Relay	× ()		It allows a periodic refresh of sessions through a re-INVITE request, to determine wheth SIP session is still active.	
	DTMF Payload Type(96~127)	101	0			
	BLF Send DTMF	Disabled	~		Pucy Lamp Fig	
	BLF DTMF Code				Busy Lamp Field/BLF List Monitors a specific extension list of extensions for status changes on IP phones.	
	Retransmission	Disabled	✓ 🕜			
	Subscribe Register	Disabled	✓ 🕜		Shared Call Ap	nearance
	Subscribe for MWI	Enabled	✓ ?		(SCA)/ Bridge Line Appearance (BLA) It allows users to share a SII on several IP phones. Any IP phone can be used to origina or receive calls on the shared line.	
	MWI Subscription Period(Seconds)	3600	0			
	Subscribe MWI To Voice Mail	Enabled	✓ ②			
	Voice Mail	*4	0			
	Voice Mail Display	Enabled	✓ 🕜		Network Conf	
	Caller ID Source	FROM	✓ Ø		It allows multipl (more than thre	

6. Click **Confirm** to accept the change.

The IP phone will subscribe to the voice mail number for MWI service using Subscribe MWI To Voice Mail.

Note MWI subscription is configurable via web user interface only.

Regulatory Notices

Service Agreements

Contact your Yealink Authorized Reseller for information about service agreements applicable to your product.

Limitations of Liability

TO THE FULL EXTENT ALLOWED BY LAW, YEALINK EXCLUDES FOR ITSELF AND ITS SUPPLIERS ANY LIABILITY, WHETHER BASED IN CONTRACT OR TORT (INCLUDING NEGLIGENCE), FOR INCIDENTAL, CONSEQUENTIAL, INDIRECT, SPECIAL, OR PUNITIVE DAMAGES OF ANY KIND, OR FOR LOSS OF REVENUE OR PROFITS, LOSS OF BUSINESS, LOSS OF INFORMATION OR DATA, OR OTHER FINANCIAL LOSS ARISING OUT OF OR IN CONNECTION WITH THE SALE, INSTALLATION, MAINTENANCE, USE, PERFORMANCE, FAILURE, OR INTERRUPTION OF ITS PRODUCTS, EVEN IF YEALINK OR ITS AUTHORIZED RESELLER HAS BEEN ADVISED OF THE POSSIBILITY OF SUCH DAMAGES, AND LIMITS ITS LIABILITY TO REPAIR, REPLACEMENT, OR REFUND OF THE PURCHASE RICE PAID, AT YEALINK'S OPTION. THIS DISCLAIMER OF LIABILITY FOR DAMAGES WILL NOT BE AFFECTED IF ANY REMEDY PROVIDED HEREIN SHALL FAIL OF ITS ESSENTIAL PURPOSE.

Safety Instructions

Save these instructions. Read these safety instructions before use!

The following basic safety precautions should always be followed to reduce the risk of fire, electrical shock, and other personal injury.

🕂 General Requirements

- Before you install and use the device, read the safety instructions carefully and observe the situation during operation.
- During the process of storage, transportation, and operation, please always keep the device dry and clean.
- During the process of storage, transportation, and operation, please avoid collision and crash of the device.
- Please do not attempt to dismantle the device by yourself. In case of any discrepancy, please contact the appointed maintenance center for repair.
- Without prior written consent, no organization or individual is permitted to make any change to the structure or the safety design of the device. Yealink is under no circumstances liable to consequences or legal issues caused by such changes.
- Please refer to the relevant laws and statutes while using the device. Legal rights of others should also be respected as well.

🔨 Environmental Requirements

- Place the device at a well-ventilated place. Do not expose the device under direct sunlight.
- Keep the device dry and free of dust.
- Place the device on a stable and level platform.

- Please do not place heavy objects on the device in case of damageand deformation caused by the heavy load.
- Keep at least 10 cm between the device and the closest object for heat dissipation.
- Do not place the device on or near any inflammable or fire-vulnerable object, such as rubber-made materials.
- Keep the device away from any heat source or bare fire, such as a candle or an electric heater.
- Keep the device away from any household appliance with a strong magnetic field or electromagnetic field, such as a microwave oven or a refrigerator.

M Operating Requirements

- Do not let a child operate the device without guidance.
- Do not let a child play with the device or any accessory in case of accidental swallowing.
- Please only use the accessories provided or authorized by the manufacturer.
- The power supply of the device must meet the requirements of the input voltage of the device. Please use the provided surge protection power socket only.
- Before plugging or unplugging any cable, make sure that your hands are completely dry.
- Do not spill liquid of any kind on the product or use the equipment near water, for example, near a bathtub, washbowl, kitchen sink, wet basement or near a swimming pool.
- Do not tread on, pull, or over-bend any cable in case of malfunction of the device.
- During a thunderstorm, stop using the device and disconnect it from the power supply. Unplug the power plug and the Asymmetric Digital Subscriber Line (ADSL) twisted pair (the radio frequency cable) to avoid lightning strike.
- If the device is left unused for a rather long time, disconnect it from the power supply and unplug the power plug.
- When there is smoke emitted from the device, or some abnormal noise or smell, disconnect the device from the power supply, and unplug the power plug immediately. Contact the specified maintenance center for repair.
- Do not insert any object into equipment slots that is not part of the product or auxiliary product.
- Before connecting a cable, connect the grounding cable of the device first. Do not disconnect the grounding cable until you have disconnected all other cables.

\Lambda Cleaning Requirements

- Before cleaning the device, stop using it and disconnect it from the power supply.
- Use a piece of soft, dry and anti-static cloth to clean the device.
- Keep the power plug clean and dry. Using a dirty or wet power plug may lead to electric shock or other perils.

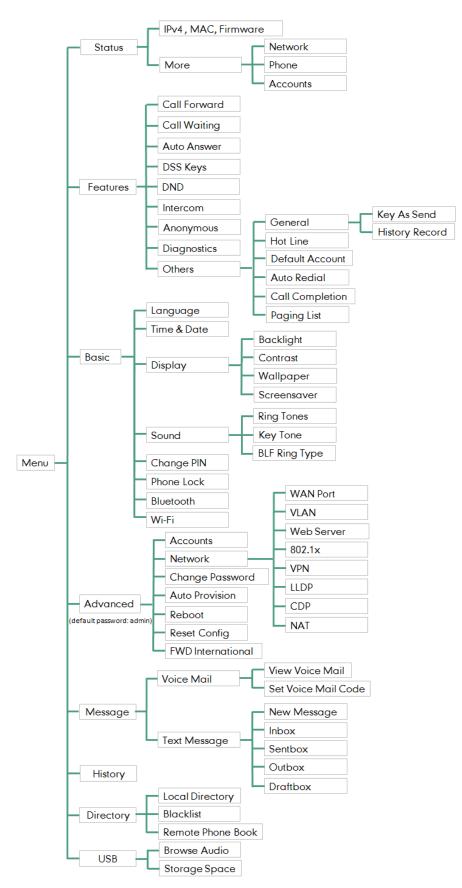
Appendix

Appendix A - Time Zones

Time Zone	Time Zone Time Zone Name	
-11	Samoa	
-10	United States-Hawaii-Aleutian, United States-Alaska-Aleutian	
-9:30	French Polynesia	
-9	United States-Alaska Time	
<u> </u>	Canada(Vancouver,Whitehorse), Mexico(Tijuana,Mexicali), United	
-8	States-Pacific Time	
7	Canada(Edmonton,Calgary), Mexico(Mazatlan,Chihuahua), United	
-7	States-MST no DST, United States-Mountain Time	
-6	Canada-Manitoba(Winnipeg), Chile(Easter Islands), Mexico(Mexico	
-0	City,Acapulco), United States-Central Time	
-5	Bahamas(Nassau), Canada(Montreal,Ottawa,Quebec), Cuba(Havana),	
	United States-Eastern Time	
-4:30	Venezuela(Caracas)	
	Canada(Halifax,Saint John), Chile(Santiago), Paraguay(Asuncion),	
-4	United Kingdom-Bermuda(Bermuda), United Kingdom(Falkland	
	Islands), Trinidad&Tobago	
-3:30	Canada-New Foundland(St.Johns)	
-3	Argentina(Buenos Aires), Brazil(DST), Brazil(no DST),	
	Denmark-Greenland(Nuuk)	
-2:30	Newfoundland and Labrador	
-2	Brazil(no DST)	
-1	Portugal(Azores)	
0	Denmark-Faroe Islands(Torshavn), GMT, Greenland, Ireland(Dublin),	
0	Morocco, Portugal(Lisboa,Porto,Funchal), Spain-Canary Islands(Las	
	Palmas), United Kingdom(London)	
	Albania(Tirane), Austria(Vienna), Belgium(Brussels), Caicos, Chad, Croatia(Zagreb), Czech Republic(Prague),	
	Denmark(Kopenhagen), France(Paris), Germany(Berlin),	
+1	Hungary(Budapest), Italy(Rome), Luxembourg(Luxembourg),	
	Macedonia(Skopje), Namibia(Windhoek), Netherlands(Amsterdam),	
	Spain(Madrid)	
	Estonia(Tallinn), Finland(Helsinki), Gaza Strip(Gaza), Greece(Athens),	
	Israel(Tel Aviv), Jordan(Amman), Latvia(Riga), Lebanon(Beirut),	
+2	Moldova(Kishinev), Romania(Bucharest), Russia(Kaliningrad),	
	Syria(Damascus), Turkey(Ankara), Ukraine(Kyiv, Odessa)	
+3	East Africa Time, Iraq(Baghdad), Russia(Moscow)	
+3:30	Iran(Teheran)	
+4	Armenia(Yerevan), Azerbaijan(Baku), Georgia(Tbilisi),	
+4	Kazakhstan(Aktau), Russia(Samara)	
+4:30	Afghanistan(Kabul)	
+5	Kazakhstan(Aqtobe), Kyrgyzstan(Bishkek), Pakistan(Islamabad),	
	Russia(Chelyabinsk)	
+5:30	India(Calcutta)	
+5:45	Nepal(Katmandu)	
+6	Kazakhstan(Astana, Almaty), Russia(Novosibirsk,Omsk)	
+6:30	Myanmar(Naypyitaw)	
+7	Russia(Krasnoyarsk), Thailand(Bangkok)	
+8	Australia(Perth), China(Beijing), Russia(Irkutsk, Ulan-Ude),	
	Singapore(Singapore)	
+8:45	Eucla	

Time Zone	Time Zone Name		
+9	Japan(Tokyo), Korea(Seoul), Russia(Yakutsk,Chita)		
+9:30	Australia(Adelaide), Australia(Darwin)		
+10	Australia(Brisbane), Australia(Hobart),		
+10	Australia(Sydney,Melboume,Canberra), Russia(Vladivostok)		
+10:30	Australia(Lord Howe Islands)		
+11	New Caledonia(Noumea), Russia(Srednekolymsk Time)		
+11:30	Norfolk Island		
+12	New Zealand(Wellington,Auckland), Russia(Kamchatka Time)		
+12:45	New Zealand(Chatham Islands)		
+13	Tonga(Nukualofa)		
+13:30	Chatham Islands		
+14	Kiribati		





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