# **About This Guide**

Thank you for choosing this Enterprise IP Phone which is especially designed for power users in the office environment. It features fashionable and sleek design, abundant telephony applications, broad interoperability with the popular 3<sup>rd</sup> party VoIP products, fulfilling the VoIP deployment needs from enterprise and ITSP.

In this User Guide, you will find everything you need to quickly use your new phone. Be sure to verify with your system administrator that your network is prepared for configuring your IP phone. As well, be sure to read the Packing List section in this guide before you set up and use the phone.

#### **Declaration of Conformity**

**CEFC** Hereby, it's declared that this phone is in conformity with the essential requirements and other relevant provisions of the CE, FCC.

#### **CE Mark Warning**

This is a class B device, in a domestic environment; this product may cause radio interference, in which case the user may be required to take adequate measures.

#### **WEEE Warning**



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.

# Contents

Getting Started	1 -
Packing List	- 1 -
Assembling the Phone	
Configuration and Registration	
Configuring via Web Page	4 -
Configuring via keypad	- 6 -
Overview	7 -
Keypad Instruction	7 -
LED Instruction	- 8 -
Icon Instruction	9 -
User Interface	- 10 -
Customizing Your Phone	13 -
General Settings	13 -
Phone Status	13 -
Language	13 -
Time and Date	14 -
Keypad Lock	16 -
Audio Settings	18 -
Volume	18 -
Ring Tones	18 -
Codec Selection	20 -
Contact Management	21 -
Edit/Add/Delete Contact	21 -
Import/Export Contact list	- 24 -
Blacklist	25 -
Remote Phone Book	- 26 -
Broadsoft	- 27 -
LDAP	- 27 -
Other Settings	- 28 -
Key as Send	- 28 -
Hot Line	- 29 -
Anonymous call	- 30 -
Auto Redial	31 -
Auto Answer	- 32 -
Missed call log	- 33 -
Broadsoft Call Log	- 33 -

Logo Customization	
Programmable Key	
Softkey Layout	
Live Dialpad	
Replace Rule	
Dial Now	- 37 -
Area Code	
Block Out	
Feature Synchronisation	
Push XML	
WatchDog	
Action URL/URI	- 43 -
Using the Basic Call Functions	- 46 -
Making a call	46 -
Call Devices	
Call Methods	46 -
Password dial	- 48 -
Call Completion	49 -
Answering a call	- 50 -
Answering an incoming call	50 -
Denying an incoming call	50 -
DND	50 -
Call Forward	50 -
Intercom	51 -
During an Active Call	52 -
Mute	52 -
Call Hold	52 -
Call Waiting	53 -
Call Transfer	53 -
3-way Conference	54 -
Network Conference	54 -
HuaWei ATS Conference	55 -
Message	57 -
Voicemail	57 -
SMS	59 -
Using the Advanced Phone Functions	- 61 -
Account Setting	61 -
TLS	
DNS-SRV	62 -
Network Setting	
PC Port Setting	
VLAN Setting	
<b>~</b>	

	LLDP	65 -
	HTTPS	66 -
Mai	ntenance Tasks	67 -
	Administrator Mode	67 -
	Administrator/User Password	67 -
	Reboot	68 -
	Reset to Factory	68 -
	Firmware Update	69 -
	Decryption	70 -
	Zero-sp-touch	71 -
	System Log Export	72 -
	PCAP Trace Export	72 -
	802.1X	73 -
DSS	Skeys Configuration	73 -
	Conference	74 -
	Forward	74 -
	Transfer	75 -
	Hold	75 -
	DND	75 -
	Redial	75 -
	Call Return	76 -
	SMS	76 -
	Pick Up	76 -
	Call Park	76 -
	DTMF	77 -
	Voice Mail	77 -
	Speed Dial	77 -
	Intercom	78 -
	Line	78 -
	BLF	78 -
	URL	79 -
	Group Listening	79 -
	Public Hold	79 -
	Private Hold	79 -
	Shared Line	80 -
	XML Group	80 -
	Group Pick up	
	Paging	81 -
	Record	
	Xml Browser	
	Hot Desking	
	URL Record	
	LDAP	
	BLF List	

Trouble Shooting	- 88 -
Ring	87 -
Voice	86 -
Tone Settings	85 -
Broadsoft Group	85 -
Local Group	84 -
ACD	84 -
Zero-sp-touch	84 -
Prefix	83 -

# **Getting Started**

# **Packing List**

The following components are included in your package:

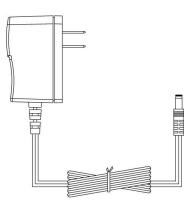
Enterprise IP Phone



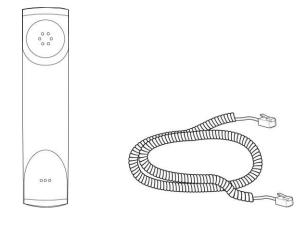
Phone Stand



• Power Adapter



• Handset & Handset Cord



Ethernet Cable



• Quick Installation Guide & Quick Reference



CD Content



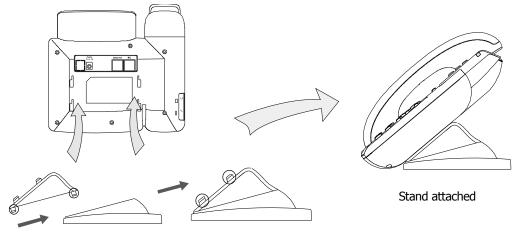
Check this list before installation to ensure that you have received each item. If you are missing any items, contact your IP phone reseller.

# **Assembling the Phone**

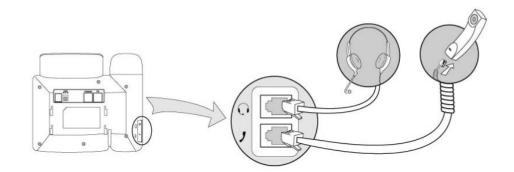
This section introduce how to assemble the phone with the components in the packing list:

- Attach the stand;
- Connect Handset and Headset;
- Connect Network and Power.

#### 1) Attach the Stand, as shown below:



2) Connect Handset and Headset, as shown below:

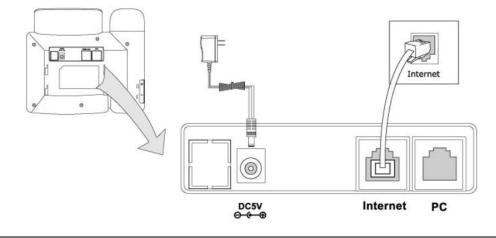


#### Note:

Headset is not provided in the packing list. Please contact your distributor for more information.

#### 3) Connect Network and Power

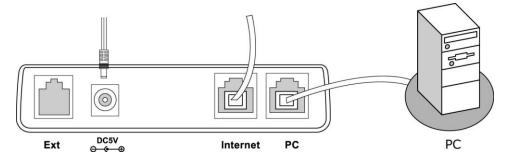
There are two ways for network and power source connections. You can either connect the phone to the AC Power directly using the power adapter or to a PoE compliant switch or hub. Your system administrator will advise you on which one to use.



#### Note:

- 1. If inline power is provided, do not install AC adapter. Make sure the Ethernet cable and switch/hub is PoE compliant.
- 2. The Internet Port can be also connected to Hub/Switch/IP PBX or other internet devices.

The phone can also share the network connection with other network devices such as PC. Connect the phone's PC port and computer's Network Port together using an Ethernet cable, shown as below:



# **Configuration and Registration**

If you are administrator, you need to do some simple configuration to make the phone work. If not, please contact your internet administrator or service provider for more details.

#### **Configuring via Web Page**

Press OK button on the keypad of the phone to enter the status page and find out the IP address of IP phone. Enter it (for example http://192.168.3.35) into the address bar of web browser. The default administrator's login name and password are **admin/admin**. The default user's login name and password are **user/user**.

#### Note:

Please locate your PC in the same network segment of IP phone (192.168.3.X) to access the web configuration page. Please consult your system administrator for help.

#### **Network Settings**

Choose Network->Internet Port (WAN) .

**DHCP:** Under the default situation the phone attempts to contact a DHCP Server in your network in order to obtain its valid network settings, e.g. IP address, sub mask, gateway, DNS server, etc.

**Static IP Address:** If your phone cannot contact a DHCP Server for any reason, you need to enter the network settings manually via Static IP Address. Please contact your internet administrator for more details.

**PPPoE:** If you are using the xDSL Modem, you can connect your phone to the internet via PPPoE mode. Please contact your ISP for the User **Name** and **Password** for internet access.

#### Note:

Using the wrong network parameters may result in inaccessibility of your phone and may also have an impact on your network performance. Please contact your network administrator.

#### Account Settings

The phone attempts to register to the SIP server using the parameters obtained by initialization.

Field	Description	
Register Status	It shows the register status of the phone.	
Account Active	You can choose on/off to enable/disable the account respectively.	
Label	The name showing on the LCD of current device.	
Display Name	The local phone name showing on the other phone when calling.	
Register Name	Register name provided by ISP.	
User Name	User account information, provided by ISP.	
Password	Account password provided by ISP.	
SIP Server	SIP server address provided by ISP.	

Choose Account, you will find the following parameters:

When you have finished the Network and Account Setting configuration, the Register Status Icons will show in the idle screen:



Registering Registered successfully Register failed

When all accounts register fail, phone will display "No Service" by default.

When the phone reboot, it will register automatically. If many phones register at the same time, this will affect the server, the users can set the register power up time so that the phone will random register automatically within the set time.

Setting the power up time via web interface:

Choose Network->Advanced-> Registration random, enter the time in the field.

#### Note:

Should the IP PBX (SIP registrar) require an authentication, you will be prompted to enter the correct password. Make sure you are using the appropriate input method or enter the password via the web user interface.

#### **Configuring via keypad**

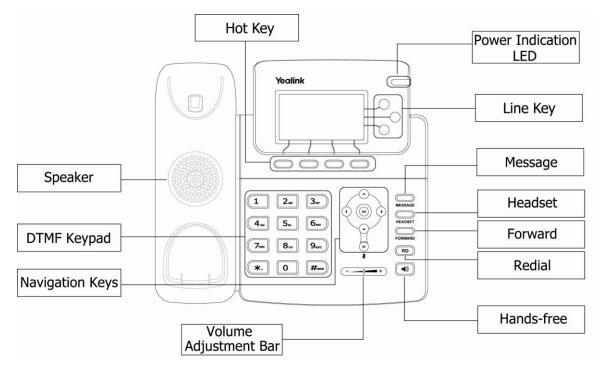
**Network Settings:** Press Menu->Settings->Advanced Settings, enter the password, and choose Network ->WAN Port /PC Port /VLAN/Webserver Type/802.1x Settings Option to enter the internet relating configuration page.

**Account Settings:** Press Menu->Settings->Advanced Settings, enter the password, and choose Accounts->Enter to configure the account settings.

You can refer to the above "Configuring via Web Page" for the parameter details.

# Overview

# **Keypad Instruction**



You can check the following list which introduces the IP phone's keypad in details:

#### **Power Indication LED**

It will show the power status, it will be on if the phone is powered, off if the phone is not powered, and blink when someone calls in or there is a call on mute.

#### Hot Key

The screen will display labels for these keys, to identify their context-sensitive functions and you can custom softkeys under different status.

#### Line Key

These buttons are used to active up to the three user accounts.

#### **DTMF Keypad**

Use the DTMF hard keys to enter numbers, letters and special characters. Depending on the selected input mode, you can enter digits, lower / upper case or special characters.

#### **Navigation Keys**

Use the navigation keys to navigate in the display menus and confirm and cancel

actions.

Cancel	Up	Down	Left	Right	Confirm
×			A B A	A B B	ОК

#### **Audio Device Control Keys**

Use the audio device control keys to perform the following actions depending on your phone type:



Adjust the volume of the handset, headset, speaker and ring tone;



Allows for hands-free communication during calls;

Press to switch to the Group Listening mode.



HEADSET: Place and receive calls through an optionally connected headset.



 $rac{1}{4}$ . Mute audio transmission locally during calls;

#### **Hard Feature Keys**



MESSAGE: Allow users to access the Voice Mail interface directly;

FORWARD: Forward the current call to third party;

When the phone is idle, press to enter the forward configuration page.

(RD): Press to enter the Dialed Calls interface, then choose a record to dial out.

# **LED** Instruction

#### Table 1 Line Keys

LED Status	Description
Steady green	The account is active
Blinking green	There is an incoming call to the account, or there is a call on hold status
off	The phone is in idle status whatever registered/ unregistered

Table 2 Line Keys set to BLF

LED Status	Description
Steady green	The monitored account is in idle status
Fast blinking green	There is an incoming call to the monitored account
Slow blinking green	The monitored account is on an conversation
Off	It is not active as BLF

Table 3 Power Indication LED

LED Status	Description
Steady green	Power on
Blinking green	There is incoming call to the device, or there is call on mute
off	Power off

Table 4 Line Keys set to BLA

LED Status	Description	
Steady green	All of the members are in idle status	
Slow blinking green	Some part(s) is seizing the line/ ring-back/ under the private hold status.	
Fast blinking green	Some part(s) is ringing /on the phone / under the public hold status or all of the parts press hold.	
Off	It is inactive as BLA	

# **Icon Instruction**

The IP Phone displays different kinds of icons on its LCD, you can refer to the following table for their meanings:

Icon	Description
Ţ	Flashes when the internet is disconnected
8	Account register failed
8	Account registering
f	Account register successful
Ľ	Missed calls
2	Call in
<b>`</b>	Call out
2aB	Input Method: all letters and numbers
123	Input Method: numbers
abc	Input Method: multi-lingual letters in lower case

ABC	Input Method: multi-lingual letters in upper case
<b>N</b>	Call mute
Φ	Call hold
00	Voicemail
$\boxtimes$	SMS
₽	Call forward
DND	DND(Do Not Disturb)
AA	Auto answer
C	In handset mode
0	In headset mode
-40	In speaker mode
⊗	The recording session cannot be started
ø	The recording cannot be stopped
Ø	Probably the recording box is full
ж	This call cannot be recorded.
•	The recording session is successfully started

# **User Interface**

There are two ways to customize specific options on your phones:

- 1. Using keypad and display on the phone;
- 2. Using Web user interface in an Internet browser from your PC; please refer to "Configuration and Registration" to get into the Web interface.

In many instances, it is possible to use both the user interfaces to operate the phone and change settings; some, however, are only possible via a phone or web user interface. Please refer to the following table for differences:

Phone Options	Phone UI	Web UI
Status		
IP		
MAC		
Firmware	$\checkmark$	$\checkmark$
Network		
Phone		
Accounts		
Call Features		

Call Forward√√Call Waiting√√DSS Keys√√Key as Send√√Hot Line√√Hot Line√√Anonymous Call√√Auto Redial Settings√√Dial Plan√√Tones√√DND Code√√ReDialTone✓√Emergency✓✓BusyToneDelay✓✓	
DSS Keys√Key as Send√Hot Line√Anonymous Call√Auto Redial Settings√Dial Plan√Tones√DND Code√ReDialTone√Emergency✓	
Key as Send√Hot Line√Anonymous Call√Auto Redial Settings√Dial Plan√Tones√DND Code√ReDialTone√Emergency√	
Hot Line√Anonymous Call√Auto Redial Settings√Dial Plan	
<ul> <li>Anonymous Call</li> <li>Auto Redial Settings</li> <li>Dial Plan</li> <li>Tones</li> <li>DND Code</li> <li>ReDialTone</li> <li>Emergency</li> </ul>	
<ul> <li>Auto Redial Settings</li> <li>Dial Plan</li> <li>Tones</li> <li>DND Code</li> <li>ReDialTone</li> <li>Emergency</li> </ul>	
Dial Plan Tones DND Code √ ReDialTone Emergency	
Tones DND Code √ ReDialTone Emergency	
DND Code√ReDialToneEmergency	
ReDialTone Emergency	
Emergency	
BusvToneDelav	
Return code when refuse	
Return code when DND	
Intercom $\checkmark$	
Call Completion √	
Basic Phone Functions	
Language $$	
Time & Date $$	
Ring Tone $\checkmark$ $\checkmark$	
Phone Volume $$	
Logo Customization $\checkmark$	
Advanced Phone Functions	
Accounts $\checkmark$	
Network $\checkmark$	
Voice	
Codecs	
Echo Cancellation etc	
Reset to factory $$	
Set admin password $$	
Update	
Set AES Key √	
Keypad Lock √	
Auto Provision $$	
Other Features	
Messages √	
History √	
Directory V	
Contacts $\checkmark$	
Blacklist √	

#### Note:

1. The above table only indicates most of phone functions rather than all of them.

Please refer to the relating parts for more details.

The default administrator password is **admin**.

# **Customizing Your Phone**

### **General Settings**

#### **Phone Status**

You can view the status of your phone using the Phone interface or the Web interface. This option allows you to review:

- Network status: IP, Mac, Gateway, DNS, WAN, LAN, etc;
- Phone status: Model, Hardware version, Firmware version, Product ID, MAC, etc;
- Accounts: The 3 SIP accounts status;
- Firmware version;

To check the Phone Status via Phone interface:

- 1) Press OK button directly or Menu hot key, choose the Status option.
- 2) Use the navigation keys to scroll through the list and check the specific one.

To view the Phone Status via Web interface:

Open the web browsers and input the IP Address http://WAN-ip-address; Enter the account and password (default account and password are both "admin"), choose Status directly to check the status.

#### Language

The default Phone interface language is **English**. The Web interface language depends on your computer Operation System. It will automatically match the language with your computer and browser.

It also supports Simplified Chinese, Traditional Chinese, French, German, Italian, Polish, Turkish, Portuguese, Spanish, etc. You can change the language for the phone user interface and the web user interface independently from each other.

#### Note:

All languages may not be available for selection. The available languages depend on the language packs currently loaded to the IP phone. Please contact with your system administrator for more information about loading language packs.

To change the language via Phone interface:

- 1) Press Menu->Settings->Basic Settings->Language.
- 2) Scroll through the list of available languages.
- 3) Press the Save hot key when the desired language is highlighted. The language appears on the graphic display will be changed to the one you chose.



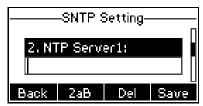
- 4) Press Save hot key to save the changes.
- 5) Press Back hot key to return to the previous screen.

### **Time and Date**

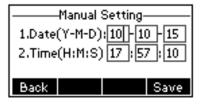
The time and date appears on the idle screen of the IP phone. If the phone cannot obtain a time and date from the call server, please contact your system administrator if the time or date is incorrect. You can set the time manually or via the SNTP server which is used to synchronize the time.

To change the Time and Date via the Phone interface:

- 1) Press Menu->Settings->Basic Settings->Time & Date.
- If SNTP Settings is chosen, the phone will automatically get the time from the specific NTP Server. Use the navigation keys to highlight the specific option and the relating changes. You can set the Time Zone, NTP Server1/Server2, and Daylight Saving respectively.



3) If Manual Setting is chosen, the time can be set manually. Use the navigation keys to highlight the option and enter the specific date and time.



4) Press Save hot key, the time appears on the idle screen will be changed. Press Back hot key to return to the previous screen.

To set the time format via the Phone interface:

1) Press Menu->Settings->Basic Settings->Time & Date->Time & Date Format.



2) Use the Switch hot key to choose a preferred time format: 12 hour or 24 hour.

- Use the Switch hot key to choose a preferred date format, the IP phone can support
   7 kind of date display format.
- 4) Press the Save hot key to save the changes and return to the previous screen.

To change the Time and Date via the Web interface:

Choose Phone->Preference to do the relating changes. You can also change the Update Interval which specifies the time frequency that the phone refreshes the time automatically. Please refer to the instruction above for the parameters' detail.

Ve elinet							<u>Logout</u>
Yealink	Status	Account	Network	Phone	Contacts	Upgrade	Security
	Preference Feat			Action URL			SMS
	Preference   Feat	tures   Surkey Lay	your   Doo Key		voice King	Torres Diar Plan	6190
	WEB Langu	age	English	× 0		D NOTE	
	DHCP Time		Disabled	~		Time	Zone
	Time Zone		+8 China	(Beijing)	*	Choos you ar	e the time zone re in.
	Primary NTP	Server	cn.pool.r	tp.org 🕜			Server
	Secondary I	NTP Server	cn.pool.r	tp.org 🕜		to syn	erver which is used ichronize the clock
	Update Inte	erval(seconds)	1000	0			phone.
	Daylight Sav	ving Time	Automat	ic 💌 🕜		Specif	te Interval y the interval at
	Fixed Type		🖲 By Da	ate 🖸 By Week			the unit will h the time.
	StartTime		Month	Day H	lour		<b>oht Saving Time</b> arameter used to
	EndTime		Month	Day H	lour		the daylight saving
	Offset(minu	tes)					al Time
	Manual Time	e	Disabled	?		Enable	e or disable to set nanually.
	Time Forma	t	24 Hour	× 0		Ring	
	Date Forma	t	WWW M	IMM DD 💌 🕜		must	pload ringtones be format of wav
	Live Dialpad		Disabled	?		should	e sampling rate I be 8K, mono, 16-
	Inter Digit 1	Fime(1~14)(seconds)	4	0		DIC U-I	aw compression
	Flash Hook	Time(<800ms)	1	0			
	Backlight Tir	me(seconds)	30	v ()			
	Keyboard Lo	ock	Disabled				
	WatchDog		Enabled	*			

To change the Daylight Saving Time settings via the Phone interface:

 Press Menu->Settings->Basic Settings->Time & Date->SNTP Settings->Daylight Saving.

	-SNTP \$	Setting–		_
4. Da	ylight S	Gaving:		
Automatic 🔸				
Back		Switch	Save	

- 2) Press the Switch hot key to choose Off/On/Automatic options.
- 3) Press the Save hot key to save the changes.

To change the Daylight Saving Time Settings via the Web interface:

- 1) Choose Phone->Preference->Daylight Saving Time to do the relating changes.
- 2) Choose Enable option, then you can set the Daylight Saving Time.
- 3) Choose Automatic. There is a table named as AutoDST.xml has been saved in the configuration file, If the table includes daylight saving time of your time zone, it will

show the Fixed Type: By Date or By Week. And the daylight saving time is unchangeable, unless to update the AutoDST.xml via auto provision.

Yealink							<u>Logout</u>
	Status	Account	Network	Phone	Contacts	Upgra	ade Security
	Preference Feat	ures 📔 Softkey La	ayout DSS Key	Action URL V	/oice Ring	Tones Di	ial Plan SMS
	WEB Langu	age	English				2 NOTE
	DHCP Time Time Zone		Disabled +8 China(	ejing)	~	0	Time Zone Choose the time zone you are in.
	Primary NTP Secondary N		cn.pool.nt	•			NTP Server The server which is used to synchronize the clock
	Update Inte Daylight Sav	erval(seconds) ring Time	1000 Automatic	0 •			of the phone. <b>Update Interval</b> Specify the interval at which the write unit
	Fixed Type		🖲 By Dat	te 🖸 By Week	Daylight Saving Tir	me settings	which the unit will resh the time.
	StartTime		Month	Day H	lour		Daylight Saving Time The parameter used to
	EndTime		Month	Day H	lour		active the daylight saving time.
	Offset(minu	tes)					Manual Time Enable or disable to set
	Manual Time	3	Disabled	<b>V</b>			Enable or disable to set time manually.
	Time Forma	t	24 Hour	✓ Ø			Ring Tone The upload ringtones
	Date Format	t	WWW MI				must be format of wav whose sampling rate
	Live Dialpad		Disabled	✓ ②			should be 8K, mono, 16- bit U-law compression
	Inter Digit T	Fime(1∼14)(seconds	;) 4	0			
	Flash Hook	Time(<800ms)	1	0			
	Backlight Tir	me(seconds)	30	?			
	Keyboard Lo	ock	Disabled	<b>• •</b>			
	WatchDog		Enabled	*			

#### Note:

By default the time zone is +8 China(Beijing), Daylight Saving Time is Automatic.

#### **Keypad Lock**

You can lock the keypad of your phone when you are temporarily not using it. This function helps you to protect your phone from unauthorized use. You can lock the following specific keys:

Menu Key: The Menu hot keys can not be used until unlocked. You can not access the menu of the phone.
Function Keys: The hard function keys (MESSAGE, FORWARD, RD, OK, X, navigation Keys, Volume adjustment button and line keys) can not be used until unlocked. You can not access the hot keys such as Conf, Hold, Mute, Tran, History, Dir, DND, etc either.
All Keys: All of the keys can not be used until unlocked. You can only use the phone to answer the incoming calls.
Lock&Answer: You can only use the phone to answer the incoming calls (But can not hung up the call by your party). Or enter the menu pages to do some configurations.

To enable keypad lock via Phone interface:

- 1) Press Menu->Settings->Advanced Settings, enter the password, and then press Confirm hot key.
- 2) Choose Keypad Lock option.
- 3) Use the navigation key to highlight the one you want to lock.
- 4) Press Save hot key to active the change, or Back hot key to return to the previous screen.
- 5) The icon  $\square$  will be displayed on the top right corner of the idle screen.
- 6) If you choose Lock&Answer, it will show the icon  $\square$  and  $\square$  on the user interface.

To unlock the phone via Phone interface:

- 1) Press Menu hot key, you are prompted for the password.
- 2) Enter the password, and then press Confirm hot key, the phone is unlocked.
- 3) The icon will be disappearing from the idle screen.
- 4) If you choose Lock&Answer, you have to enter Menu->Settings->Advanced Settings->Keypad Lock to disable this option.

To enable keypad lock via Web interface:

Choose Phone->Preference-> Keyboard Lock to do the relating changes. Please refer to the instruction above for the parameters' detail.

Yealink							<u>Logout</u>
	Status	Account	Network	Phone	Contacts	Upgrade	Security
P	Preference Feat	ures Softkey L	ayout 🛛 DSS Key	Action URL V	oice Ring T	ones 🛛 Dial Plan	SMS
	WEB Langu: DHCP Time Time Zone Primary NTP Secondary I Update Inte Daylight Sav Fixed Type StartTime EndTime Offset(minu Manual Time Time Forma Date Forma Live Dialpad Inter Digit T Flash Hook	age Server ITP Server erval(seconds) ring Time tes) t t Time(1~14)(second Time(<800ms) me(seconds)	English Disabled +8 China( cn.pool.nt 1000 Automati © By Dal Month Disabled 24 Hour WWW MI	Beijing) p.org ? p.org ? p.org ? ? P.org ? ? P.org ? ? ? ? ? ? ? ? ? ? ? ? ? ?		<ul> <li>NOTE</li> <li>Time Z</li> <li>Choose you are to sync of the to sync of the</li> <li>Updati Specify which is verified to sync of the</li> <li>Updati Specify which is verified to sync of the sync must b whose should</li> </ul>	one the time zone in. erver which is used hronze the clock phone. e Interval the interval at the unit will the time. ht Saving Time rameter used to the daylight saving

#### Note:

1. The default password for unlock is **admin**.

### **Audio Settings**

#### Volume

You can adjust the volume of handset/speaker/headset/Ring Volume.

To adjust the volume when you are not in an active call:

- 1) Press Menu->Settings->Basic Settings.
- 2) Scroll to Phone Volume, and press Enter hot key, highlight the one you want to adjust the volume, use the Volume Adjustment Bar or navigation keys to adjust the volume.

Phone Volume						
1. Handset Volume						
2. Speaker Volume						
3. Headset Volume						
Back Enter						

- 3) Press OK hot key to save the change or Back hot key to cancel.
- 4) And you can also press the Volume Adjustment Bar to adjust the ring volume when the phone is in idle status.

To adjust the volume when you are on an active call:

When Handset/Headset/Hands-free mode is activated, press the Volume Adjustment Bar to a comfortable level.

#### Note:

The volume can only be adjusted via Phone interface. When you adjust the ring volume to 0, or press the Silence hot key when there is incoming call, the icon  $\mathbb{Z}^X$  will be shown on the LCD. Press the Volume Adjustment Bar to adjust the volume, and the icon will disappear.

#### **Ring Tones**

You can adjust the type and volume of the ring tone.

To adjust the Ring Tone Type via Phone interface:

- 1) Press Menu->Settings->Basic Settings.
- 2) Scroll to Ring Tone, and press Enter hot key.
- 3) Use the navigation keys to highlight the specific one.



4) Press Save hot key to save the change or Back hot key to cancel.

To change the Ring Tone Type via Web interface:

#### **Using the Basic Call Functions**

Choose Phone->Preference->Ring Type, highlight the specific one in the scroll-down menu, click confirm button to update the change. You can also delete the specific one by clicking the Delete button.

Time Zone		+8 China(Beijing)	~	0	you are in.
Primary NTI	P Server	cn.pool.ntp.org	0		NTP Server
Secondary	NTP Server	cn.pool.ntp.org	0		The server which is used to synchronize the clock
Update Int	erval(seconds)	1000	0		of the phone.
Daylight Sa	ving Time	Automatic 💌	0		Update Interval Specify the interval at
Fixed Type		🖲 By Date 🖸 By We	eek		which the unit will refresh the time.
StartTime		Month Day	Hour		Daylight Saving Time The parameter used to
EndTime		Month Day	Hour		active the daylight saving time.
Offset(minu	utes)				Manual Time
Manual Tim	e	Disabled 💌	0		Enable or disable to set time manually.
Time Forma	at	24 Hour 💌	0		Ring Tone
Date Forma	t	WWW MMM DD 💌	0		The upload ringtones must be format of wav whose sampling rate
Live Dialpad		Disabled 💌	0		should be 8K, mono, 16- bit U-law compression
Inter Digit	Time(1~14)(seconds)	4	0		
Flash Hook	Time(<800ms)	1	0		
Backlight T	ime(seconds)	30 💌	0		
Keyboard L	ock	Disabled 💌	0		
WatchDog		Enabled 💌			
Ring Type		Ring1.wav 💌	Del 🕜		
Upload Ring	jtone	Ring1.wav Ring2.wav Ring3.wav	浏览		
		Ring4.wav	el		
	Confirm	Ring6.wav Ring7.wav			
	Committee	Ring8.wav		_	
Note:					

The ring tone file of system cannot be deleted.

To upload the new Ring Tone via Web interface:

- 1) Choose Phone->Preference->Upload Ringtone.
- 2) Click Browse button to choose the specific ring tone file.
- 3) Click Upload button to upload the file.

#### Note:

- 1. The ring tone file format must be in 16bits WAV format (via Ulaw Compression), 8K sample rate (monophony).
- 2. Blank or other special characters can not be included in the file name.

To specify ring tones for a specific account via Web interface:

Choose Account->Basic->Ring Type option, and highlight the preferred one for the chosen account in the scroll-down menu, then click confirm button to save the changes.

#### **Using the Basic Call Functions**

	різріаў напіс	101	<b>V</b>	User account, provided by
	Register Name	101	0	VoIP service provider.
	User Name	101	0	NAT Traversal Defines the STUN server will
	Password	•••	0	be active or not.
	SIP Server	10.1.4.47	Port 5060 🕜	Proxy Require
	Enable Outbound Proxy Server	Disabled	× 0	A special parameter just for Nortel server. If you login to
	Outbound Proxy Server		Port 5060 🕜	Nortel server, the value shou be:
	Transport	UDP	× 0	com.nortelnetworks.firewall
	Backup Outbound Proxy Server		Port 5060 🕜	Codecs
	NAT Traversal	Disabled	× 🕜	Choose the codecs you wan to use.
	STUN Server		Port 3478 🕜	Advanced
	Voice Mail		0	The Advanced parameters for administrator.
	Proxy Require		0	administrator.
	Anonymous Call	Off	✓ Ø	
	On Code			
	Off Code		0	
	Anonymous Call Rejection	Off	× 0	
	On Code		0	
	Off Code		0	
	Missed call log	Enabled	✓ Ø	
	Auto Answer	Disabled	× 0	
	Ring Type		<u> </u>	
Codecs >>	0	common Ring1.wav		
Codecs >>	V	Ring2.wav Ring3.wav		
Advanced >>		Ring4.wav Ring5.wav Ring6.wav Ring7.wav		

#### **Codec Selection**

The IP phone supports the following voice codecs: G722, G723\_53, G723\_63, G726-16, G726-24, G726-32, G726-40, G729, PCMU and PCMA.

You can enable/disable the desired codecs via Web interface. Please contact your System Administrator for more details about the codecs.

To enable/disable the codecs:

1) Choose Account->Codecs.

Yealin	ık		-		-		<u>Loqout</u>
Easy Vop	Status	Account	Network	Phone	Contacts	Upgrade	Security
A	Account		Account	:1 💌		D NOTE	
В	Basic >>					SIP ser	<b>y Name</b> vice subscriber's name will be used for Caller ID
c	Codecs >> 🕜	Disable codecs 6723_53 6723_63 6726-16 6726-24 6726-32 6726-40	Enable PCMU PCMU PCMA G729 G722 <<	codecs ↑ ↓		SIP ser used fr User N User a VOIP s NAT T Define be act Proxy A spec	ccount, provided by ervice provider. raversal s the STUN server will we or not. Require ali parameter just for
A	Advanced >>	Confirm		Cancel		Nortel be: com.nu Codec Choose to use Advan	e the codecs you want dvanced parameters for

- 2) Use the navigation keys to highlight the desired one in the Enable/Disable Codecs list, and press the >> / << to move to the other list.
- 3) Click Confirm to save the change.

#### Note:

Codec Selection can only be set via Web interface.

# **Contact Management**

#### **Edit/Add/Delete Contact**

You can store a large number of contacts in your phone's contact list. You can add, edit, delete, dial, and search for a contact in the contact list.

The directory includes Local Directory, Blacklist , Remote Phonebook and Broadsoft.

To add a Group via Phone interface:

- 1) Press Dir->Local Directory
- 2) Press Group hot key to enter to the Add Group page.
- 3) Enter the group name and choose the ring.
- 4) Press the save hot key to save.

Add Group						
Nam	e:					
Back	abc	Delete Save				

To add a contact via Phone interface:

- 1) Press Dir->Local Directory
- 2) Choose a group, and press the Enter hot key.
- Press Add hot key, enter Name, Office, Mobile, Other and Account options of the contact by the keypad. Use the abc hot key to select between numeric and upper/lower case alphanumeric modes.
- 4) Use the navigation keys to select the desired account as Line if you want to assign the contact to a specific account.
- 5) Choose and set a special ring tone for the contact.
- 6) Use the navigation keys to select the group which you want to assign.
- 7) Press Save hot key to add the contact or Back hot key to cancel the change.

•	Cor	ntact	_1/1 <b>}</b>
<b>4</b> 152			
Back	Add	Option	Dial

To edit a contact via Phone interface:

- 1) Press Dir->Local Directory
- 2) Choose a group, and press the Enter hot key.
- 3) Use the navigation key to highlight the one you want to edit, press Option->Detail, enter to the edit interface.
- 4) And then make the changes, press Save hot key to save the changes, or press Back hot key to return to the directory.

	1	52	
Nam	le;		
152			
Back	abc	Delete	Save

To delete a contact via Phone interface:

- 1) Press Dir->Local Directory
- 2) Choose a group, and press the Enter hot key.
- 3) Use the navigation key to highlight the one you want to delete, press Option hot key, and scroll to Delete, press OK hot key.
- 4) It will pop up a warning frame asking whether confirm to delete the contact.
- 5) Press OK hot key to confirm the operation, or press the Cancel hot key to return to the directory.

_ · · · · · · · · · · · · · · · · · · ·	itact1/1▶						
hwwwwwmMau	ningwwwwww						
Delete Selected Item?							
[Cancel]	[Ok]						
Cancel	Ok						

To move a contact to the Blacklist via Phone interface:

- 1) Press Dir->Local Directory
- 2) Choose a group, and press the Enter hot key.
- 3) Use the navigation key to highlight the one you want to move, press Option hot key, and scroll to Move to Blacklist, press OK hot key.
- 4) It will pop up a warning frame asking whether confirm to move the contact.
- 5) Press OK hot key to confirm the operation, or press the Cancel hot key to return to the directory

	taot1/1
	ning~~~~~
Move selected	to blacklist ?
[Cancel]	[0k]
Cancel	Ok

#### Note:

If the contact is moved to the blacklist, then the call from this contact can not get through.

To move a contact in History to Contacts via Phone interface:

- 1) Press History hot key to enter the call history list.
- 2) Use the navigation keys to highlight a record, and then press the Option hot key to pop up the frame, highlight Add to Contacts option, then press the OK hot key to enter the edit interface.
- 3) Press the abc hot key to switch the input mode.
- 4) After the edition, press the Save hot key to save the change. Then you can go to Contacts interface to check the record.

Search Contact									
Search:									
Cancel	abc	Delete	Ok						

To search a contact via Phone interface:

- 1) Press Dir->Local Directory->Search.
- 2) It will turn to the search interface, and then you can enter the query condition, press the OK hot key.
- 3) Then the phone will show the record which qualified.

S	earch	Contact _	
Searc	h:		
Cancel	abc	Delete	Ok

To add/delete/edit/move the contacts via the Web interface:

Choose Contacts->Local Phone Book. Please refer to the instruction above for the parameters' detail.

							<u>Logout</u>
Yealink	Status	Account	Network	Phone	Contacts	Upgrade	Security
	Local Pho	oneBook BlackList	Remote PhoneBo	ok   Phone Call Inf	o LDAP E	Broadsoft Cal	Log
Contacts Index 1 2 3 4	All Contacts Name aa ss vvv	Office Num N 123456 0599123456 5896	lobile Num Othe	Num Account Auto Auto Auto		Fill Fill sh	DTE Id Contact/Blacklist out the contact ormation. User ouldn't leave contact me blank.
5 6 7 8 9 10 Page: 1	▼ Prev	Next	Move To BlackLi	t	Delete All	Co Se Wa Gri bu	elete ntact/Blacklist lect the contact you int to delete in the d, and then press the tton Delete to nfirm.
<b>Con</b> Nam Offic Mob	tacts			Information Auto	Velete All	Mi Cc Ch wa gri bu Co co it.	ove to ntact/Blacklist oose the contacts you int to move in the d, and press the tton move to ntact/Blaklist to move
Acco Ring Grou	ps [	Auto  Auto Auto Auto Auto Auto Auto Auto Auto	Impo	t XML Export XM	浏览 L 浏览	Bri for Clin cri na	iport owse the file in XML mat. port ck Export button and aste a file with whose me you prefer to port.

### **Import/Export Contact list**

Import/Export Contact List via Web interface:

1) Choose Contacts-> Local Phone Book.

#### **Using the Basic Call Functions**

Vea	link	-				-	-	<u>Loqout</u>
EBS	VOP-	Status	Account	Network	Phone	Contacts	Upgrade	Security
		Local Phon	eBook BlackList	Remote PhoneBoo	k   Phone Call Info	LDAP B	roadsoft 📔 Call Lo	g
	Index 1 2 3 4 5 6 7 8	All Contacts Name aa ss vvv	♥ ⑦ Office Num Mo 123456 0599123456 5896	ble Num Other I	Num Account Auto Auto Auto	<u>Ha</u> Groups	Fill ou Fill ou should name	Contact/Blacklist t the contact nation. User dn't leave contact blank. te act/Blacklist t the contact you to delete in the
	9 10 Page: 1 Cont	acts	Next		formation	Delete All	Del Del Del Del	e to cact/Blacklist se the contacts you to move in the
	Name Office Mobile Other Accou	Num Num Num	uto 💌	Groups Ring Add		elete All	butto Conta it.	and press the in move to act/Blaklist to move ort se the file in XML
	Ring Group		uto 💙 I/A 💙 Jit Search	Import	XML Export XML	刘览	forma Expo Click f creat	rt Export button and e a file with whose you prefer to
				Import	CSV Export CSV			

- 2) Browse the specific contact list file in .XML or .csv format, and then click Import button. The imported contact lists will be showed in the Contact List.
- 3) Move the mouse to the icon ?, you will see the notes for parameters. The meanings of this icon on other pages is the same, we will not elaborate it one by one.
- 4) Click the Export button to export the contact list.

#### Note:

Import/Export Contact List can be only set via Web interface.

### Blacklist

If you add a contact to blacklist, then the call from this contact cannot get through. The operation of blacklist is the same as contact.

Status     Account     Network     Phone     Contacts     Upgrade       Local PhoneBook     BlackList     Remote PhoneBook     Phone Call Info     LDAP     Broadsoft     Call Loc       BlackList     Imdex     Name     Office Num     Mobile Num     Other Num     Account     Imdex       Index     Name     Office Num     Mobile Num     Other Num     Account     Imdex     Imdex       1     abc     124785     0147852     Auto     Imdex     Imdex     Imdex       3     Imdex     Imdex     Imdex     Imdex     Imdex     Imdex     Imdex       3     Imdex     Imdex     Imdex     Imdex     Imdex     Imdex     Imdex       4     Imdex     Imdex     Imdex     Imdex     Imdex     Imdex     Imdex       9     Imdex     Imdex     Imdex     Imdex     Imdex     Imdex     Imdex       Name     Im	Yeal	link						<b>`</b>	Lo	<u>qout</u>
BlackList       Panque         Index       Name       Office Num       Mobile Num       Other Num       Account       I       Image: Control of Contacts       Image: Contacts	Easy		Status	Account	Network	Phone	Contacts	Upgrade	Security	
Index       Name       Office Num       Mobile Num       Other Num       Account       I         1       abc       124785       0147852       Auto       I       blackle			Local PhoneBook	BlackList	Remote PhoneBoo	k   Phone Call Ir	nfo LDAP B	roadsoft   Call Lo	g	
Index     Name     Office Num     Mobile Num     Other Num     Account       1     abc     124785     0147852     Auto     □       3     □     □     □       4     □     □       5     □     □       6     □     □       7     □     □       8     □     □       9     □     □       10     □     □       Page:     1     Prev     Next       Name     □     □       Office Num     □		BlackList	0				Ha	angup	r	
1     auc     1247632     Auto       2     7       3     7       6     7       7     7       8     7       9     7       10     7       Page:     1       Prev     Next       Move To Contacts     Delete All       Delete Num		Index	Name	Office Num	Mobile Num	Other Num	Account			
3 4 5 6 7 8 9 10 Page: 1 ♥ Prev Next Move To Contacts Delete All Del Name Office Num		1	abc	<u>124785</u>	0147852		Auto	D black	listnote	
4		2								
5     □       6     □       7     □       8     □       9     □       10     □       Page:     1 ♥ Prev       Name     □       Office Num     □		3								
6 7 8 9 10 Page: 1 V Prev Next Move To Contacts Delete All Del Name Office Num		4								
7 8 9 10 Page: 1  Prev Next Move To Contacts Delete All Del Name Office Num		5								
8 9 10 Page: 1  Prev Next Move To Contacts Delete All Del Name Office Num		6								
9 10 Page: 1  Prev Next Move To Contacts Delete All Del Name Office Num		7								
10 Page: 1  Prev Next Move To Contacts Delete All Del Name Office Num		8								
Page:     I     Prev     Next     Move To Contacts     Delete All     Del       Name		9								
Name Office Num		10								
Name Office Num		Page: 1 💌	Prev	Next	Move To Contact	:S	Delete All	Del		
Other Num Account Auto		Office N Mobile N Other N Account	lum um Auto							

### **Remote Phone Book**

The IP phone has directory itself, but in the enterprise applications where there are a need for a common phone book. For the maintenance and the update of it, the common phone book is usually carried out on the server or IPPBX to maintain up-to-date public phone book, terminal users need to have remote phone book function. When the users browse the remote phone book, the terminal will check and download the latest information released on the server in time, and display on the terminal for the user.

To set the Remote Phone Book via Web interface:

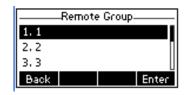
1) Choose Contacts->Remote Phone Book.

Yealir	ak 🕹							<u>Logout</u>
		Status	Account	Network	Phone	Contacts	Upgrade	Security
		Local PhoneBook	BlackList	Remote PhoneBook	Phone Call Info	LDAP Br	oadsoft 📔 Call Log	
	Index 1 2 3 4 5	http://adsca	Phone Boo	sk Url	Phone abd	Book Name	This fea to dowr from the the pho	e phone book ture allows you iload contact list s server. Input nebook URL and the phonebook

2) Input the Phone book URL and the phone book name, then click the Confirm button to save the changes.

To check the Remote Phone Book via Phone interface:

- 1) Press the Dir->Remote phonebook.
- Enter to the Remote Group page, choose a special one, and press the Enter hot key, it will go to the corresponding URL address to download the contact information for you.



#### Note:

- 1. This IP phone can support 5 remote phone books at most.
- 2. Every contact in the remote phone book can set several phone numbers.

### **Broadsoft**

Broadsoft phone book is the same as the Remote Phone Book. This feature allows you to download contact list from the server. Input the phonebook URL, port, username, password and rename the phonebook

Exact Vote       Status       Account       Network       Phone       Contacts       Upgrade       Security         Local PhoneBook       BlackList       Remote PhoneBook       Phone Cal Info       LDAP       Broadsoft       Cal Log         Broadsoft Item       Item1 V       ?       Pone       Contacts       Upgrade       Security         Broadsoft Item       Item1 V       ?       Pone       Contacts       UDAP       Broadsoft       Cal Log         Broadsoft Item       Item1 V       ?       Pone       Contacts       UDAP       Broadsoft       Cal Log         Broadsoft Item       Item1 V       ?       Pone       Contacts       UDAP       Broadsoft       Cal Log         Broadsoft Item       Item1 V       ?       Pone       Contacts       Encode       Broadsoft       Encode       Broadsoft       Encode       Broadsoft       Encode       Encode	Yealink							<u>Loqou</u>	t
Broadsoft Item Item1 V V Displayname 111 V Server 111 V User 111 V Password ··· V V V Broadsoft Directory This feature allows you to download contact list from the server. Input the phonebook URL port, username, password and rename the phonebook		Status	Account	Network	Phone	Contacts	Upgrade	Security	
Biodsoft term     Remi       Displayname     111       O     This feature allows you       Server     111       Port     111       O     This feature allows you       User     111       Password     Image: Content of the server. Input       User     111       O     Password		Local PhoneBo	ok BlackList	Remote PhoneBool	k Phone Call Info	LDAP Broa	adsoft   Call Log		
		Displayname Server Port User	111 111 111 111 111 111 000		0		Broadso This feat to down from the the phor URL,port and rena	ture allows you load contact list server. Input nebook tusername,password ame the	

### LDAP

LDAP can support these functions:

1. Search the contact: Press the DSSkey which is set up as LDAP, input a number or
letter in the new interface, the phone will search the contact in LDAP server which
following the Certain rules, and show it in the LCD, user can choose the contact to call
out.

#### **Using the Basic Call Functions**

2. Search the incoming call: the phone will search the local directory when there is a coming call. If they can't find the contact in the local directory, it will search them through LDAP server, and show the contact name in the LCD. The LDAP Lookup For Incoming Call option can be configured to enable or disable this function via web interface.

3. The function of Dial-up directory: under the dial interface, each time you press a key there are inquiring for a number. It will show on the LCD and let the user to choose. The LDAP Lookup For PreDial/Dial option can be configured to enable or disable this function via web interface.

To set the LDAP via the Web interface:

- 1) Choose Contacts->LDAP.
- 2) Configure the corresponding options
- 3) Click Confirm to save the change.

Yealink							<u>Loqout</u>
	Status	Account	Network	Phone	Contacts	Upgrade	Security
							Security
	Local PhoneBo	ok BlackList Re	mote PhoneBook	Phone Call Info	D LDAP Broa	idsoft Call Log	
	LDAP	Name Filter		0	)	NOTE	
	LDAP	Number Filter		0		LDAP s	ettings
	Serve	r Address	0.0.0.0	•			
	Port		389	0	1		
	Base			0	)		
	UserN	ame		0	)		
	Passw	ord		0	)		
	Max. H	Hits(1~32000)	50	()	)		
	LDAP	Name Attributes		0	)		
	LDAP	Number Attributes			)		
	LDAP	Display Name			1		
	Proto	col	Version3	× 0			
	Searc	h Delay(ms)(0~2000)	0	•	1		
	LDAP	Lookup For Incoming C	all Enabled	· · · · · · · · · · · · · · · · · · ·			
	LDAP	Sorting Results	Enabled	?			
	LDAP	Lookup For PreDial/Dial	Disabled	· · · · · · · · · · · · · · · · · · ·			
		Confirm		Canad			
		Commit		Cancel			

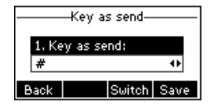
### **Other Settings**

#### Key as Send

Users can set a specific button ("#" or "\*") to active as the send button.

To set the send key via the IP phone interface:

1) Press Menu->Features->Key as Send->Enter to enter the configuration page.



- Press the Switch hot key to choose a button that you want to use as the send key:
   "#", "\*", or disable this option.
- 3) Press the Save hot key to save the changes.

To set the send key via the Web interface:

- 1) Choose Phone-> Features-> Key As Send.
- 2) Highlight the specific one in the pull-down menu, then click the Confirm button to save the change.

Yealink				Logout
Status	Account	Network Phone	Contacts	Upgrade Security
Preference Feat	tures   Softkey Layout	DSS Key Action URL	Voice Ring Tones	Dial Plan SMS
Forward: Always Targ On C Off ( Busy Targ	et Gode Code	Con Off	0 0 0	NOTE Forward This feature allows you to forward an incoming call to another phone number. Target The number to which the incoming calls will be forwarded.
On C Off C No Ans	Code		0	On Code The code that will be sent to PBX when it is switched On.
Afte Targ On C			0 0 0	Off Code The code that will be sent to PBX when it is switched Off.
Off ( General Ir Call Wai	nformation:	Enabled	0	Call Waiting This call feature allows your phone to accept other incoming calls during the conversation.
Call Wai Auto re	ting Tone dial	Enabled 💌 Disabled 💌	0 0	Key As Send Select * or # as the send key.
	# in User Name	#    Enabled	? Set keys as call out	Hotline Number When you pick up the e, it will dial out the e number automatically.
Button Send Si		Enabled 💌		Upload Logo The picture must be

### **Hot Line**

To set the hot line number via the IP phone interface:

1) Press Menu->Features->Hot Line->Enter to enter the configuration page.

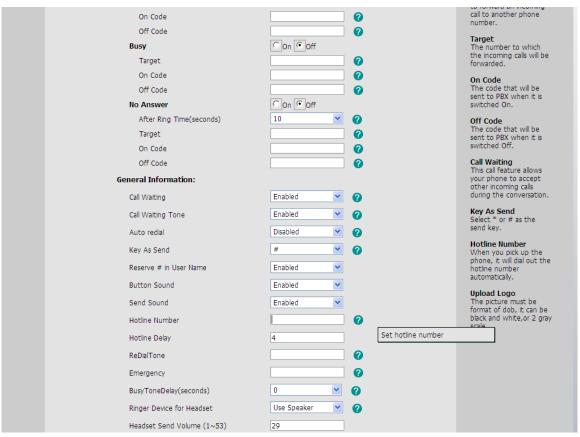
Hot Line			
1. Number:			
Back	123	Del	Save

2) Enter the hot line number and HotLine Delay time (for example, 20 seconds), then press the Save hot key to save the changes.

3) When you pick up the handset or press the speaker button, and it will dial out the number automatically if you do not press any keys for 20 seconds.

To set the Hot Line via the Web interface:

- 1) Choose Phone-> Features.
- 2) Input the Hotline Number and Hotline Delay, then click the Confirm button to save the change.



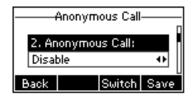
#### **Anonymous call**

To set the anonymous call via the IP phone interface:

1) Press Menu ->Features-> Anonymous Call ->Enter to enter the configuration page.

Anonymous Call		
1. Line ID:		
Line 1	<u> </u>	
Back	Switch Save	

- 2) By the Switch hot key, you can choose the Line ID.
- Press the navigation keys to enter and choose whether to enable the anonymous call function. This feature allows the subscriber to make a call with the display of their calling identification information blocked.



4) If you want to realize the anonymous call by server, please choose and enter the Call On Code and Call Off Code. When you choose to enable the anonymous call function on your IP phone, it will send information to the server, and the server will enable/ disable the anonymous call function for your IP phone automatically.

Anonymous Call			
3. Call On Code:			
Back	123	Del	Save

5) Press the navigation keys to enter and choose whether to enable the anonymous rejection function. The feature allows the subscriber to reject all calls from callers who have blocked the display of their calling identification information (calling number and calling name).

	Anonym	ous Call	
5. Rejection: Disable			-
Back		Switch	Save

6) If you want to realize the anonymous rejection by server, please choose and enter the Reject On Code and Reject Off Code. When you choose to enable the anonymous reject function on your IP phone, it will send information to the server, and the server will enable/ disable the anonymous reject function for your IP phone automatically.

#### Note:

This configuration is only available for the current default account.

7) Press the Save hot key to save the changes.

To set the anonymous call via the Web interface:

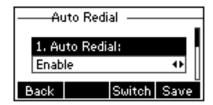
- 1) Choose Account-> Basic-> Anonymous Call to do the relating changes. Please refer to the instruction above for the parameters' detail.
- 2) Then click the Confirm button to save the changes.

#### **Auto Redial**

Auto redial is a telephone feature that redials a busy number in a fixed number of times before giving up.

To set auto redial via the IP phone interface:

1) Press Menu->Features->Auto Redial ->Enter to enter the configuration page.



- 2) By the Switch hot key, you can choose whether to enable the auto redial function.
- Press the navigation keys to choose and set the redial interval, which is used to the set period to redial again when the call is denied access or expire. It is measured by seconds.
- 4) Press the navigation keys to choose and set the redial times.
- 5) Press the Save hot key to save the changes.

To set auto redial via the Web interface:

- 1) Choose Phone->Features->Auto Redial.
- 2) Choose Enabled or Disabled in the pull-down menu, click confirm button to save the change.

Yealink					<u>Logout</u>
Statu	s Account	Network	Phone	Contacts	Upgrade Security
Preference	Features Softkey La	yout DSS Key	Action URL Void	ce Ring Tones	Dial Plan SMS
	vard:   Vways Target On Code Off Code Busy Target On Code		Coff C		NOTE Forward This feature allows you to forward an incoming call to another phone number. Target The number to which the incoming calls will be forwarded.
,	On Code Off Code Io Answer	C On	(7     (7		On Code The code that will be sent to PBX when it is switched On.
	After Ring Time(seconds) Target On Code	10	· · · · · · · · · · · · · · · · · · ·		Off Code The code that will be sent to PBX when it is switched Off.
	Off Code eral Information: Call Waiting	Enable	ed 🕑 🕜		Call Waiting This call feature allows your phone to accept other incoming calls during the conversation.
	Call Waiting Tone	Enable			Key As Send Select * or # as the send key.
	(ey As Send teserve # in User Name	# Enable		Enable/Disable auto-	redial n you pick up the phone, it will dial out the hotline number automatically.
	Button Sound Gend Sound	Enable			Upload Logo The picture must be

### **Auto Answer**

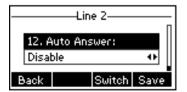
Auto-answer allows an incoming call to be answered without requiring any action by the user. This is a useful feature for people who have difficulty in using their hands or fingers, who have a visual impairment, or who have a cognitive impairment. You can set this function to a special account.

To set Auto Answer via the IP phone interface:

1) Press Menu->Settings->Advanced Settings, enter the password and press enter

hot key.

- 2) Then choose Accounts->Line X to enter the configuration page, use the navigation keys to choose Auto Answer option.
- 3) Press the Switch hot key to enable or disable the auto answer function. The default is Disable.
- 4) Press the Save hot key to save the changes.



To set Auto Answer via Web interface:

- 1) Choose Account->Basic->Auto Answer.
- 2) Choose Enabled or Disabled in the pull-down menu, click confirm button to save the change.

### Missed call log

Defines whether to save the missed calls to the call history record or display the missed calls on the idle screen. This function can only be set via the Web interface:

- 1) Choose Account->Basic->Missed call log.
- 2) Choose Enabled or Disabled in the pull-down menu, click confirm button to save the change.

# **Broadsoft Call Log**

This feature allows you to download call history from the server. Input the URL, port, username, password and rename the phonebook

Configure Call Log via web interface:

- 1) Choose Contacts->Call Log
- 2) Choose the Calllog Item.
- 3) Input the display name, URL, port, username and password.
- 4) Click the confirm button to save the changes.

#### **Using the Basic Call Functions**

Yealink	-		-			-	<u>Loqout</u>
ESSY VOP	Status	Account	Network	Phone	Contacts	Upgrade	Security
	Local PhoneBook	BlackList	Remote PhoneBook	Phone Call Info	LDAP Broa	dsoft   Call Log	
9	Callog Item DisplayName Server Port User Password	Call Log1 V 111 http://1111. 111 ? 111 •••		Cancel		to down from the the	ure allows you load call history : server. Input :,username,password ime the

Check Call Log via phone interface:

- 1) Press Menu->History Type->Network Call Log
- 2) Choose the Call log Item, press Enter softkey to download the call log.

### Logo Customization

You can upload your own logo which displayed on the idle screen.

1) Choose Phone-> Features-> Use Logo->Custom Logo via the Web interface, press Browse button, and choose the corresponding file.

Intercom Mute	Disabled 💌	0	
Intercom Tone	Enabled 💌	0	
Intercom Barge	Enabled 💌	0	
Call Completion	Disabled 💌	0	
Enable Semi-Attend Transfer	Enabled 💌	0	
Blind Transfer OnHook	Enabled 💌		
Attend Trans OnHook	Enabled 💌		
Transfer on Conference Hang up	Disabled 💌		
Feature Key Synchronisation	Disabled 💌		
Time Out for Dial-now Rule	1		
ACD Auto Available	Disabled 💌		
ACD Auto Available Timer(0~120s)	60		
RFC 2543 Hold	Disabled 💌		
Use Outbound Proxy In Dialog	Enabled 💌		
IsDeal180	Enabled 💌		
Logon Wizard	Disabled 💌		
PswPrefix			
PswLength	0		
PswDial	Disabled 💌		
PushXML Server IP			
Use Logo	Custom Logo 🔽	Del 🕜	
Upload Logo (The pixel < 132*64)	Upload Canc	浏览	ogo settings,you can select System logo,you can also choose Custom logo to upload your own ogo

2) Press Upload button to complete the Logo Customization. You will find the desired logo display on the idle screen.

#### Note:

- 1. You can also upload the Logo by AUTO PROVISION.
- 2. By the Web interface, users can set a logo to be a System Logo or a Custom Logo, and the Custom Logo can be deleted.
- 3. Only support DOB format file, For more information please contact the administrator.

### **Programmable Key**

The hot-key, navigation keys and function keys on the keypad are editable. Users can customize specific features for these keys according to their actual needs.

This function can only be set via the Web interface:

1) Choose Phone->DSS Key->Programmable Key.

	Status Accor	unt Network	Pho	ne Contacts	Upgrade Securi
Prefe	erence Features So	ftkey Layout   DSS K	iey Action U	RL Voice Ring	Tones Dial Plan SMS
Line Key	(>> <b>(</b> )				NOTE
Program	nmable Key >> (				
Key	Туре	Line		Extension	
SoftKe	y1 History	Auto	×		
SoftKe	y2 Directory	Auto	×		
SoftKe	History	Auto	~		5
SoftKe	Menu	Auto			
Up	SMS NewSMS	Auto			-
	Forward		 		
Dowr	Call Return	Auto			
Left	XML Group	Auto			
Right	Status	Auto	×		
OK	Speed Dial Local Group	Auto	×		
Cance	Broadsoft Group	Auto			
FWD	Broadsoft PhoneBook	Auto	×		

- 2) Choose and customize specific features for these keys.
- 3) Click Confirm button to save the change.

### Softkey Layout

The phone can support 12 kinds of call interface to set up softkey, user can setup different function key according to his/her own requirement or habit.

To set up softkey via web interface:

1) Choose Phone->Softkey Layout.

- 2) In the "Custom Softkey" field, chose Enable in the pull-down menu.
- 3) You can choose the corresponding call states which you want to set up the softkey in the Call States field.
- 4) Highlight the desired one in the Unselect Softkeys/Select Softkeys list, and press the 
   ↓ ← to move to the other list.
- 5) And you can use 1 / 1 to choose the order how to display in the call states.
- 6) Click confirm to save the changes.
- 7) You can also click the Reset to Default button to reset the softkeys interface.

Yealink							<u>Loqout</u>
	Status	Account	Network	Phone	Contacts	Upgrade	Security
Pre	Cu Cal Un Et Un C Li Li Li	stom SoftKey I States selected Softkeys mpty istory irectory all Switch ne Selection bool	ayout   DSS Key Disabled Dialing Selected S (ordered b Send IME Delete Cancel ←	oftkeys y position)** t	oice   Ring   T	Tones Dial Plan  Tones Dial Plan  Tones  Send  It will b  when t  Switcd  It will b  when t  Line  It will b  when t  than tu  accoun  Only CG the cas	SMS e available only there is a number e available only there are not less wo active calls e available only there are not less wo available
	Cont	firm C	ancel	Reset to Default			

#### Note:

1. We can add the Empty key more than once, but others can only choose once time, and can't be repeat. when you set up the Selected Softkeys key more than 4 options, the LCD of phone will transfer the forth key with "More" automatically, which can be use to switch to the next page, and continue to show up the other key.

# Live Dialpad

Defines whether to dial out the dialed number automatically.

This function can only be set via the Web interface:

- 1) Choose Phone->Preference->Live Dialpad.
- 2) Enable or disable it in the pull-down menu.
- 3) Click Confirm button to save the change.

# **Replace Rule**

A dial plan establishes the expected number and pattern of digits for a telephone number. This includes country codes, access codes, area codes and all combinations of digits dialed. For example if you set the *Prefix* as 0 and *Replace* as 0086 (Chinese country code), when you dial 05702000 out, the number will be replaced by 00865702000 automatically.

To set a Dial Plan via the Web interface:

1) Choose Phone->Dial Plan->Replace Rule.

Yealink									<u>Loqout</u>
	Status		Account	Network	Phone	Conta	cts	Upgrade	Security
	Preference	Features	Softkey Layout	DSS Key	Action URL	Voice Ring	Tones	Dial Plan	SMS
Repl	lace Rule >>	0						D NOTE	
In	ndex	Prefix		Replace		Account		Digit 0	-9 *
	1	0		12456789		1		Identifie	es a specific digit
	2	1		05991234					use # if it is as send key).
	3							[digit-c	liniti
	4							Identifie	any digit dialed
	5							that is in range.	ncluded in the
	6							-	
	7								<b>ligit,digit]</b> s a range as a
	9						Ē		separated list.
	10						Ē	x	
Pre	efix		Replace		Account[				s any single aracter which is
	_		Add	Edit D	el				s an arbitrary
Dial-	-now>> 🕜							number	of digits.
Area	a Code>>								
Bloc	k Out>> 🕜								

- 2) Enter the desired *Prefix*, *Replace* and *Account*.
- 3) Press Add button to save the changes.
- 4) You can also delete a specific one from the dial plan list.
- 5) You can select a record to modify, then click Edit button to submit.

### **Dial Now**

Dial-now enables you to define the specific length of any number/letter in advance(for example xxx), next time when users dial out the 123 whose length matches the Dial-now rule, the phone will dial out 123 immediately without pressing Send button.

To set a Dial Plan via the Web interface:

1) Choose Phone->Dial Plan->Dial Now.

#### **Using the Basic Call Functions**

Yealink	-			-		<u>Loqout</u>
Easy vor Status	Account	Network	Phone	Contacts	Upgrade	Security
Preference Features	Softkey Layout	DSS Key	Action URL Voice	Ring Tones	Dial Plan	SMS
Index         1         2         3         4         5         6         6	Dial-now Rule 0599		Account 1		(do not defined [digit-o Identifie that is i range. [digit-o	es a specific digit use # if it is as send key).
7 8 9 10					comma <b>x</b> Matche	s a lange as a separated list. s any single aracter which is
Dial-now Rule Area Code>> Block Out>>	Add	Account				s an arbitrary of digits.

- 2) Enter the number in Dial-now Rule and Account.
- 3) Press Add button to save the changes.
- 4) You can select a record to modify, then click Edit button to submit.
- 5) You can also delete a specific one from the dial plan list by pressing Del button.

#### Note:

1. If need to replace the unknown contents, then you can use (.) or (x), "." stand for a string of char, "x" stand for any one char. The content in () stand for a variable, the first variable is expressed by 1, the second variable is expressed by 2, the rest can be done in the same manner. For example: if you want to replace the any input content with the content beginning with 8. Input (.) in Prefix box, and input 8\$1 in Replace box.

### Area Code

Area codes are also known as Numbering Plan Areas (NPAs). These are necessary (for the most part) only when dialed from outside the code area and from mobile phones. Area codes usually indicate geographical areas within one country, although the correlation to geographical area is becoming obsolete. For non-geographical numbers, as well as mobile telephones outside of the United States and Canada, the "area code" does not correlate to a particular geographic area.

To add the area code via the Web interface:

1) Choose Phone->Dial Plan->Area Code.

Yealink						<u>Loqout</u>
Status	Account	Network	Phone	Contacts	Upgrade	Security
Preference F	eatures Softkey Lay	out   DSS Key	Action URL Void	ce   Ring   Tor	nes Dial Plan	SMS
	e .ength(1-15) Length(1-15)	0599 1 15 2 Cance	<u>d</u>		(do not defined [digit- Identifi that is i range. [digit- Specifie comma X Matche digit/ch dialed.	es a specific digit : use # if it is as send key).

2) Enter the Code and Account, set the Min Length and the Max Length option, and then click the Confirm button to save.

### **Block Out**

The specific phone numbers can be forbidden to be call out from your IP phone.

- 1) Choose Phone->Dial Plan->Block Out.
- 2) Enter the phone number and Account, and then click Add button to save the changes, or choose the specific one in the list, click Delete button to delete the record.
- 3) You can select a record to modify, then click Edit button to submit.
- 4) You can not dial out the number from your IP phone unless it is removed from the Forbidden List.

Yealink							Logout
Easy vop	Status	Account	Network	Phone	Contacts	Upgrade	Security
	Preference Features	Softkey Layout	DSS Key	Action URL Voice	Ring Tones	Dial Plan	SMS
Dial- Area Bloc	lace Rule >> ? now>> ? a Code>> ck Out>> ? Index	Block Out Number		Account		(do not defined [digit-c	es a specific digit use # if it is as send key). ligit]
	1 2 3 4 5 6 7 8 9 10	059914785 12345		Account		that is ir range. [digit-c Specifie comma X Matches digit/ch dialed.	is any digit dialed included in the <b>ligit,digit]</b> s a range as a separated list. s any single aracter which is s an arbitrary
	Block Out Number	Add	Acco Edit Del				of digits.

### Note:

1. In the Account field, you can enter 1,2,3..., "1" represents Account 1, "2" represents Account 2 ....., if the account box is empty, it mean this rule works for all accounts .

### **Feature Synchronisation**

When enabled the synchronize function, configure the DND/FWD function on device or server, DND/FWD status on device and server will be in correspondence.

To set Feature Key Synchronisation via the Web interface:

- 1) Choose Phone->Features-> Feature Key Synchronisation
- 2) There is a pull-down menu in the Type field, choose Intercom from the list.
- 3) Choose whether to enable this function from the pull-down menu.
- 4) Click the Confirm to save the change.

### **Using the Basic Call Functions**

Hotline Number		0	phone, it will dial out the hotline number automatically.
Hotline Delay	4		Upload Logo
ReDialTone		0	The picture must be format of dob, it can be
Emergency		0	black and white,or 2 gray scale.
BusyToneDelay(seconds)	0 💌	0	
Ringer Device for Headset	Use Speaker 💌	0	
Return code when refuse	486 (Busy here)	✓ ②	
Return code when DND	480 (Temporarily not a	wailable) 🔽 🕜	
DND On Code		0	
DND Off Code		0	
Enable Semi-Attend Transfer	Enabled 💌	0	
Allow Intercom	Enabled 💌	0	
Intercom Mute	Disabled 💌	0	
Intercom Tone	Enabled 💌	0	
Intercom Barge	Enabled 💌	0	
Call Completion	Disabled 💌	0	
Attend Trans OnHook	Enabled 💌		
Feature Key Synchronisation	Disabled 💌		
Use Logo	Disabled Enabled	0	
Confirm	Cancel		

## Push XML

Users configure the server's IP address on Web page, when phone receives request, it will determine whether to display corresponding content on the phone which sent by the specified server or not.

To set Push XML via Web interface:

- 1) Choose phone->features-> PushXML Server IP, enter the server IP in the field.
- 2) Click the confirm button to save the change.

#### **Using the Basic Call Functions**

Intercom Mute	Disabled	~	0
Intercom Tone	Enabled	*	0
Intercom Barge	Enabled	*	0
Call Completion	Disabled	*	0
Enable Semi-Attend Transfer	Enabled	*	0
Blind Transfer OnHook	Enabled	*	
Attend Trans OnHook	Enabled	*	
Transfer on Conference Hang up	Disabled	*	
Feature Key Synchronisation	Disabled	*	
Time Out for Dial-now Rule	1		
ACD Auto Available	Disabled	*	
ACD Auto Available Timer(0~120s)	60		
RFC 2543 Hold	Disabled	*	
Use Outbound Proxy In Dialog	Enabled	*	
IsDeal180	Enabled	*	
Logon Wizard	Disabled	*	
PswPrefix			
PswLength	0		
PswDial	Disabled	*	
PushXML Server IP			
Use Logo	System Logo	*	0

# WatchDog

When 'WatchDog' function is 'Enabled', phone will auto reboot after 10 seconds if some important process of phone crash. When 'Disable' the function , the phone will not reboot.

Configure watchdog via web interface:

Choose phone-> Preference->WatchDog, in the pull-down menu, choose enable or disable this function.

### **Using the Basic Call Functions**

Secondary NTP Server Update Interval(seconds) Daylight Saving Time Fixed Type	cn.pool.ntp.org	The server which is used to synchronize the clock of the phone. Update Interval Specify the interval at which the unit will refresh the time.
StartTime EndTime Offset(minutes) Manual Time Time Format Date Format Live Dialpad Inter Digit Time(1~14)(seconds) Flash Hook Time(<800ms) Backlight Brightness Backlight Time(seconds) LCD Contrast Keyboard Lock WatchDog Ring Type Upload Ringtone	Month Day   Hour   Month   Day   Hour   Disabled   VWWW MMM DD   VWWW MMM DD   VOUL   Disabled   V   VWWW MMM DD   VOUL   VOUL   Disabled   VOUL   Disabled   VOUL   Participation   VOUL   VO	Daylight Saving Time The parameter used to active the daylight saving time. Manual Time Enable or disable to set time manually. Ring Tone The upload ringtones must be format of wav whose sampling rate should be 8K, mono, 16- bit U-law compression

## **Action URL/URI**

Action URL: Record the operation of phone, send these corresponding information to server; Action URI: Remote control phone for corresponding operation.

The operation can be recorded include: Setup Completed, Log On, Log Off, Register Failed, Off hook, On hook...etc.

Set Action URL via web interface:

- 1) Choose Phone->Action URL
- 2) Enter the Corresponding contents
- 3) Click Confirm to save the changes.

Yealink						<u>Loqout</u>
Status	Account	Network	Phone	Contacts	Upgrade	Security
Preference Features So	ftkey Layout	DSS Key EXT Ke	y Action URL	Voice Ring	Tones Dial P	lan SMS
Setup Completed Log On Log Off Register Failed Off hook On hook Incoming call Outgoing call Call established Call terminated Open DND Close DND Open Always Forward Close Always Forward Open Busy Forward Close Busy Forward Open No Answer Forward					■ NOTE ActionUR	

### Action URI:

Enter the "http://<u>Phone ip</u>/cgi-bin/ConfigManApp.com?<u>key=xxx</u>" in Browser address bar, the phone will realizing the corresponding function. If you not login the web with the user name and password, you will need to specify the user/password to confirm the operation. The username/password can be added into the URI like: http://<u>admin:admin@10.2.3.25</u>/cgi-bin/ConfigManApp.com?<u>key=OK</u>

Our phone can support the following functions function:

Phone ip stand for the phone's IP address. Key=xxx stand for the following rules:

To answer the call: key=OK/key=ENTER To turn on speaker mode: key=SPEAKER Press transfer button: key=F\_TRANSFER Increasing the volume: key=VOLUME\_UP Reduce the volume: key=VOLUME\_DOWN To mute the call: key=MUTE To hold the call: key=F\_HOLD To end the call: key=F\_HOLD To enter the DTMF number(include Numeric , \* or # keys): key=0-9/\*/POUND Press a line key: key=L1-L6 Press Conference button: key=F\_CONFERENCE Press a soft key: key=F1-F4 Press Message button: key=MSG

Press Headset button: key=HEADSET Press RD button: key=RD Press navigation key: key=UP/ DOWN/ LEFT/ RIGHT To reboot the phone: key=Reboot To check the Auto provision: key=AutoP To enable DND: key=DNDOn To disable DND: key=DNDOff

Dial out:

http://<u>phone IP</u>/cgi-bin/ConfigManApp.com?number=<u>NUMBER</u>&outgoing\_uri=<u>URI</u> Phone ip stand for the phone's IP address. NUMBER stand for the number which you want to send. URI stand for the account.

For example:

http://<u>10.2.3.25</u>/cgi-bin/ConfigManApp.com?number=<u>0599123456</u>&outgoing\_uri=<u>21</u> <u>6@192.168.1.199</u>

NOTE:

If there is no account, or the account abnormal, the phone will dial out with the default account.

# **Using the Basic Call Functions**

# Making a call

### **Call Devices**

You can make a phone call via the following devices:

- 1) Pick up the handset, **C** icon will be showed in the idle screen.
- 2) Press the Speaker button, 🖤 icon will be showed in the idle screen.
- 3) Press the Headset button if the headset is connected to the Headset Port in advance. The icon **O** will be showed in the idle screen.

You can also dial the number first, and then choose the method you will use to speak to the other party.

### **Call Methods**

You can dial the number directly if you didn't register a account but fill the SIP Server in the registered interface. But the number which you dial must be in the same SIP server.

You can press an available line button if there is more than one account, then

- 1) Dial the number you want to call, or
- 2) In dial-up interface, press Pool hot key, use the navigation button to highlight your choice, enter into the corresponding option or
- Press the RD button to enter the Dialed Calls interface, choose a record to call out, or
- 4) Press the DSS keys which have been set as speed dial button.

Then press the Send button or Send hot key to make the call out if necessary.

And you can also dial-up via web interface:

1) Choose Contact->Local Phone Book/BlackList, click the number which you want to dial out, and then the phone will dial out by default account.

#### **Using the Basic Call Functions**

Yealink					Loqout
	atus Account	Network	Phone Conta	acts Upgra	ide Security
Lo	cal PhoneBook   Black	List Remote PhoneBook	Phone Call Info LD	AP Broadsoft	Call Log
Contacts All Cont	tacts 💌 🕜			Hangup	
Index Nan	ne Office Num	Mobile Num Other Nu	um Account (	Groups 🗌	D NOTE
1 aa	a <u>123456</u>		Auto		Add Contact/Blacklist
2 55	<u>0599123456</u>		Auto		Fill out the contact information. User
3 vv	v <u>5896</u>		Auto		shouldn't leave contact
4					name blank.
5					
6					Delete
7					Contact/Blacklist Select the contact you
8					want to delete in the
9					grid, and then press the button Delete to
10					confirm.
Page: 1 💙	Prev	Move To BlackList	Dele	te All Del	
Contacts		Group Infe			Move to Contact/Blacklist
Contacts		Group Inte	ormation		Choose the contacts you want to move in the
Name		Groups			grid, and press the
Office Num		Ring	Auto	~	button move to Contact/Blaklist to move
Mobile Num					it.
Other Num		Add	Edit Del Delete Al		
	Auto III	]			Import
Account	Auto 🗸	Please sele	ect the contacts list file		Browse the file in XML format.
Ring	Auto		浏览		rormat.
Groups	N/A 🚩				F
		Import X	ML Export XML		Export Click Export button and
Add	d Edit Search		浏览	-	create a file with whose
			以成		name you prefer to export.
		Import C	SV Export CSV		
		Import e			
		Show ti	itle		

- 2) Or choose Contact->Phone Call Info, enter the number in the Dial a Number, select the line from the Outgoing Identity list. Then click the dial button to call out.
- 3) Or choose Contact->Phone Call Info, click the number which you want to dial out from the call list, the phone will dial out by corresponding account.

#### **Using the Basic Call Functions**

Yealink				_			<u>Logout</u>
Easy vop	Status	Account	Network	Phone	Contacts	Upgrade	Security
	Local Pho	neBook BlackList	Remote PhoneBo	ok Phone Call Info	D LDAP B	roadsoft Ca	all Log
Contact	s All Contacts				Hai	ngup	
Index	Name	Office Num Mo	bile Num Other	Num Account	Groups	E •	IOTE
1	qq	<u>123</u>		Auto			dd Contact/Blacklist
2	WW	<u>0599</u>		Auto			ill out the contact formation. User
3						sł	nouldn't leave contact
4							ame blank.
6						÷.	elete
7						C C	ontact/Blacklist
8							elect the contact you ant to delete in the
9						۲ g	rid, and then press the
10							utton Delete to onfirm.
Page: 1	Y Prev	Next	Move To BlackLis	t	Delete All	Del	
Cor	itacts		Group I	nformation		C	Iove to ontact/Blacklist hoose the contacts you rant to move in the
Nan	ne 🗌		Groups			g	rid, and press the
Offi	ce Num		Ring	Auto	*		utton move to ontact/Blaklist to move
Mob	oile Num					it	
Oth	er Num		Add	Edit Del C	Delete All		
Acc	ount 🛛	Auto 🔽					<b>mport</b> rowse the file in XML
Ring	, 7	Auto 💌	Please s	elect the contacts I			ormat.
Gro	ups I	N/A 🔽			浏览		
	Add	idit Search	Import	ı	L	C ci n	<b>xport</b> lick Export button and reate a file with whose ame you prefer to xport.
			Import	t CSV Export CSV	V		

4) You can click the Hangup button to end the call in the web page.

### **Password dial**

When number entered is beginning with the password prefix, the following N numbers after the password prefix will be hidden as \*, N stand for the value which you enter in the PswLength field. For example: you set the password prefix is 3, enter the PswLength is 2, then you enter the number 34567, it will display 3\*\*67 on the phone. Set password dial via web interface:

- 1) Choose phone->features->PswDial, in the pull-down menu, choose enable.
- 2) Enter the password prefix in the PswPrefix field
- 3) Enter the Length in the PswLength field.
- 4) Click the confirm to save the changes.

Intercom Mute	Disabled 💌	0	
Intercom Tone	Enabled 💌	0	
Intercom Barge	Enabled 💌	0	
Call Completion	Disabled 💌	0	
Enable Semi-Attend Transfer	Enabled 💌	0	
Blind Transfer OnHook	Enabled 💌		
Attend Trans OnHook	Enabled 💌		
Transfer on Conference Hang up	Disabled 💌		
Feature Key Synchronisation	Disabled 💌		
Time Out for Dial-now Rule	1		
ACD Auto Available	Disabled 💌		
ACD Auto Available Timer(0~120s)	60		
RFC 2543 Hold	Disabled 💌		
Use Outbound Proxy In Dialog	Enabled 💌		
IsDeal180	Enabled 💌		
Logon Wizard	Disabled 💌		
PswPrefix			
PswLength	0		
PswDial	Disabled 🕑		
PushXML Server IP			
Use Logo	System Logo 🛛 👻	0	

# **Call Completion**

Have encountered such a situation? When you call a contact, but the other side is busy on a call. Do you want the server to inform you immediately when the contact end the call, in order to establish a conversation with each other in time? Call Completion can help you to solve this problem.

To configure Call Completion via phone interface:

1) Press the following hot keys: Menu->Features ->Call Completion to enter the configuration page.

	Call Compl	etion	
	ll Complet	ion:	4
Disat			•
Back	Su	Jitch	Save

- 2) By the Switch soft key, choose whether to enable this option.
- 3) Press the Save soft key to save your changes.

# Answering a call

### Answering an incoming call

- 1) If you are not on another phone, lift the handset using, or press the Speaker button/ Answer hot key to answer using the speakerphone, or press the headset button to answer using the headset.
- 2) If you are on another call, press Answer hot key to answer the call, or press the Reject hot key to refuse it.
- 3) You can also press the Silence hot key to mute the call.

During the conversation, you can alternate between Headset, Handset and Speaker phone by pressing the corresponding buttons or picking up the handset.

### **Denying an incoming call**

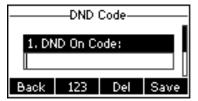
Press Reject hot key to deny the incoming call directly.

### DND

Press DND hot key to active DND Mode when the phone is in idle status. Further incoming calls will be rejected and the display shows: End icon. Press it again to deactivate DND mode. You can find the incoming call record in the Call History.

You can also set DND function by the DND Code:

1) Go to Menu->Features->DND Code to enter the configuration interface.



- 2) Set the DND On Code and the DND Off Code, then press the Save hot key to save the changes.
- 3) When you press the DND hot key, the phone will send a message to the server, and the server will turn on the DND function. Then any calls to the extension will be rejected by the server automatically. And the incoming call record will not be displayed in the Call History.

# **Call Forward**

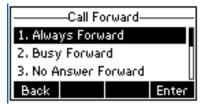
This feature allows you to forward an incoming call to another phone number e.g. a cell phone or voice mailbox.

The following call forwarding events can be configured:

- *Always:* Incoming calls are immediately forwarded.
- *Busy:* Incoming calls are immediately forwarded when the phone is busy.
- *No Answer:* Incoming calls are forwarded when the phone is not answered after a specific period.

To configure Call Forward via Phone interface:

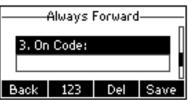
1) Press the following hot keys: Menu->Features->Call Forward->Enter.



- 2) There are 3 options: Always Forward, Busy Forward and No Answer Forward.
- 3) Choose one of them, enter the phone number you want to forward.

———Always I	Forwar	dр
2. Forward to 3444444	):	
Back 123	Del	Save

4) If you want to realize this function by the server, please enter the On Code and Off Code option, then when you choose to enable the call forward function via your IP phone, it will send message to the server, and the server will turn on the function immediately. When there are calls to the extension, the server will forward it to the set number automatically based on the forward type. And the IP phone will not show the record in the call history anymore.



5) Press Save to save the changes.

To configure Call Forward via Web interface:

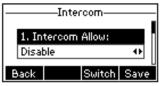
Choose Phone->Features->Call Forward to do the relating changes. Please refer the above configuration information.

### Intercom

Intercom mode is useful in an office environment as a quick access to connect to the operator or the secretary.

To configure Intercom option via phone interface:

1) Press the following hot keys: Menu->Features->Enter->Intercom->OK to enter the configuration page.



- 2) Intercom Allow: To set whether to answer the incoming intercom calls.
- 3) Intercom Mute: To set whether to mute the incoming intercom calls automatically.
- 4) Intercom Tone: To set whether to play ring tones when there is incoming intercom calls to your extension.
- 5) Intercom Barge: To set whether to answer the incoming intercom calls during a conversation. If the option is enabled, when there is incoming intercom calls to your extension, if you are on an intercom conversation, it will refuse the call automatically; or it will put the current call on hold and put the incoming intercom call through.
- 6) Choose and set the different options by navigation keys and the Switch soft key.
- 7) Press the Save soft key to save your changes.

# **During an Active Call**

### Mute

This function allows you to mute the microphone of the active audio device during a call; you can not be heard by the other party. You can still hear all other parties while mute is enabled.

To mute/resume the conversation:

Press MUTE button during a conversation to mute all the conversations, the icon will be shown on the LCD. Press it again to get the microphone return to normal conversation. When you press the MUTE button all of the conversation will be muted.

### Call Hold

This call function allows you to place an active call on hold. In this case your IP PBX might play a melody or message to the other party while waiting. Other calls can be received and made while having a call on hold.

To hold/resume a call:

- 1) Press the HOLD button or Hold hot key to put your active call on hold.
- 2) If there is only one call on hold, press the Resume hot key or Hold button to retrieve the call.
- 3) If there are more than one call on hold, press the line keys to switch the current account (The on hold calls are under different lines), or the Up/Down button to highlight the call (The on hold calls are under the same lines), then press the Resume hot key or Hold button to retrieve the call.

Note:

When you are under the call hold status, putting down the handset, the conversation

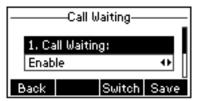
will go on over the speaker instead of hanging up the call.

### **Call Waiting**

This call feature allows your phone to accept other incoming calls to the extension no matter under which circumstances.

To enable/disable Call Waiting via Phone interface:

- 1) Press Menu->Features->Call Waiting hot keys.
- 2) Use the navigation keys or Disable/Enable hot key to active/inactive call waiting.



- 3) Use the navigation keys or Switch hot key to enable/disable the Play Tone option. This option used to define whether to play ring tones when there is call incoming during an active call.
- 4) Press Save hot keys to save the changes, or Back hot key to return to the previous menu.

To enable/disable Call Waiting via Web interface:

Choose Phone->Features->Call Waiting to do the relating changes.

### **Call Transfer**

You can customize your phone so that incoming calls are transferred directly to the third party such as another extension, mobile phone number, etc. There are three ways to transfer the call: Blind Transfer, Attended Transfer and Semi-Attended Transfer.

To Blind Transfer via phone interface:

- A and B is on an conversation, A press TRAN Button or Tran hot key to put B on hold, then A can dial the third telephone number C (or press Directory soft key to enter the contact list, then choose a record) and press the Tran hot key to call out. A will turn to the hold status, and the LCD will display as Transferred.
- 2) After the C answered it, or A press the Cancel hot key to complete the transfer.
- 3) A will be disconnected from the call. B can talk to C.
- 4) If C refused to answer the call, it will prompt A that the transfer operation is failed. If the current mode is speaker, it will ring up; if the current mode is handset or headset, it will play ring tones for every five seconds. Pressing any function keys to exit the prompt interface. This function should be supported by server.

To Attended Transfer via phone interface:

 A and B is on an conversation, A press TRAN Button or Tran hot key to put B on hold, then A can dial the third telephone number C (or press Directory soft key to enter the contact list, then choose a record) and press the OK or SEND button to call out.

- 2) After C answered it, A and C can have a private conversation without B hearing it, then A press the Tran button to complete the transfer.
- 3) A will be disconnected from the call. B can talk to C.

To Semi-Attended Transfer via phone interface:

- 1) A and B is on an conversation, A press the TRAN button or Transfer hot key to put B on hold, then A can dial a new number C (or press Directory soft key to enter the contact list, then choose a record) and press the OK or SEND button to call out.
- 2) While C is ringing, A hang up or press the Tran hot key. Then A will turn to the hold status, and the LCD will display as Transferred.
- 3) You will be disconnected from the call, when C pick up, B can talk to C.

#### Note:

Make sure that the SIP server you have registered supports this function.

### **3-way Conference**

You can establish a three-party conference, during the conversation three phone parties can communicate with each other.

To establish a conference:

- 1) Press the Conf hot key during an active call.
- 2) The first call is placed on hold. You will hear a dial tone. Dial the number to conference in (or press Directory soft key to enter the contact list, then choose a contact to conference in), then press the Send hot key.
- 3) When the call is answered, you can have a private conversation at first. And then press the CONF button, the conference call will include you and the other two parties.
- 4) During the conference, press the Split hot key to split the conference into two hold lines, and press the Resume hot key to resume the chosen call respectively.
- 5) When you press the hold key, the conference will be on hold.
- 6) Hang up to disconnect all parties.

### **Network Conference**

If you want to make a conference with more than three people, you can open the function of network conference.

If you enabled this function, you can put the meeting conference on the server.

To enable network conference via web interface:

1) Choose Account->Account X->Advance->Conference Type, there is a pull-down menu, choose network from the list.

#### **Using the Basic Call Functions**

Subscribe Register	Disabled 💌	0
Subscribe for MWI	Disabled 💌	0
MWI Subscription Period(Scope:0~84600) (seconds)	3600	
Caller ID Header	FROM 💌	0
Use Session Timer	Disabled 💌	0
Session Timer(seconds)		0
Refresher	Uac 💌	0
Use user=phone	Disabled 💌	0
Voice Encryption (SRTP)	🖸 On 🖲 Off 🕜	
ptime(ms)	20 💌	0
BLF List URI		0
BLF List Code	*97	
Shared Line	Disabled 💌	0
Dialog-Info Call Pickup	Disabled 💌	0
BLA Number		0
BLA Subscription Period(Scope:60~7200)	300	0
SIP Send MAC	Disabled 💌	0
SIP Send Line	Disabled 💌	0
SIP Registration Retry Timer(Scope:0~1800) (seconds)	30	0
Enable Signal Encode	Disabled 💌	
Signal Encode Key		
Conference Type	Local 💌	
Conference URI	Local ATS	
ACD Subscription Period(120~3600)	Network	
Confirm	Cancel	

- 2) Enter the Conference URI.
- 3) Press Confirm button to save the changes.

To establish a conference:

- 1) Press the Conf hot key during an active call.
- 2) Dial the number to conference in, then press the Send hot key
- 3) When the call is answered, press the CONF button.
- 4) After starting a three way conference, press Conf button to enter Conference dialing interface and invite another party to participate in teleconference.
- 5) After starting conference, press Hold key to Hold local call without influencing others in conference.

# HuaWei ATS Conference

Turn on this function can make Multi-Party Conference come true; you can add or delete any attenders, also you can have a private chat with any member. This function needs the server's support.

To enable ATS conference via web interface:

 Choose Account->Account X->Advance->Conference Type, there is a pull-down menu, choose ATS from the list.

### **Using the Basic Call Functions**

		_
Subscribe Register	Disabled 💌	0
Subscribe for MWI	Disabled 💌	0
MWI Subscription Period(Scope:0~84600) (seconds)	3600	
Caller ID Header	FROM 💌	0
Use Session Timer	Disabled 💌	0
Session Timer(seconds)		0
Refresher	Uac 💌	0
Use user=phone	Disabled 💌	0
Voice Encryption (SRTP)	🖸 On 💽 Off 🛛 🕜	
ptime(ms)	20 💌	0
BLF List URI		0
BLF List Code	*97	
Shared Line	Disabled 💌	0
Dialog-Info Call Pickup	Disabled 💌	0
BLA Number		0
BLA Subscription Period(Scope:60~7200)	300	0
SIP Send MAC	Disabled 💌	0
SIP Send Line	Disabled 💌	0
SIP Registration Retry Timer(Scope:0~1800) (seconds)	30	0
Enable Signal Encode	Disabled 💌	
Signal Encode Key		
Conference Type	Local 💌	
Conference URI	Local ATS	
ACD Subscription Period(120~3600)	Network	
Confirm	Cancel	

- 2) Enter the Conference URI.
- 3) Press Confirm button to save the changes.

To establish a conference:

- 1) Press the Conf hot key during an active call.
- 2) Dial the number to conference in, then press the Send hot key
- 3) When the call is answered, press the CONF button.
- 4) After the CONF is established, initiator can continually add CONF members, press Add hot key to enter into dial-up interface, enter the number, press send button to dial out, when the people answer the call, he have been attended to the CONF.
- 5) Initiator can use private chart function to have a private chart with any member, press PriChat, choose any members in phone member list, and then press PriChat hot key, both sides enter into private chart mode, if there only 3 CONF members, both sides begin to private chart and the third party enter a Hold mode. If the attender is more than 3, and then the other attenders keep on talking. If you want to end the private chat, press ExitPri hot key.
- 6) Initiator can use remove function to remove any CONF members, press remove hot key, choose any members in phone member list , and then press remove hot key again.
- 7) Initiator Hangs up, the CONF over.

# Message

The phone supports SMS (Short Messaging Service) and Voicemail, if you want to use them, please make sure that your VoIP telephony system supports this functionalities and your accounts' message has been enabled.

# Voicemail

Your voice mailbox messages, which are usually stored on a media server of your local or hosted VoIP telephony system, can be accessed from your phone.

New voice messages can be indicated both acoustically and visually as described below:

- The idle screen will indicate the new voice messages coming:
- The MESSAGE button will be lighted.

To configure the Voicemail code via Phone interface:

- 1) Press Menu->Messages->Voice Mail->Set Voice Mail .
- 2) Use the navigation keys to highlight the line for which you want to set, enter the code which the phone uses to connect to your system. Press 123 hot key to choose the proper input method.
- 3) Press Save to save the change, press Back to return to the previous menu.

#### Note:

Please contact your system administrator for the connecting code. Different systems have different codes.

Want to see amount of Voice mail via phone interface, must enable the Subscribe for MWI via the web interface at first.

- 1) Choose Account->Advanced-> Subscribe for MWI.
- 2) Choose enable in the pull-down menu.

### **Using the Basic Call Functions**

Codecs >>       Register Name         Advanced >>       UDP Keep-alive Message       Enabled       Image: Codecs and the subscriber's ID         UDP Keep-alive Interval(seconds)       30       Image: Codecs and the subscriber's ID         UDP Keep-alive Interval(seconds)       30       Image: Codecs and the subscriber's ID         Logn Expre(seconds)       300       Image: Codecs and the subscriber's ID         RPort       Disabled       Image: Codecs and the subscriber's ID         SIP Session Timer(seconds)       1       0.5       Image: Codecs and the subscriber's ID         SIP Session Timer(seconds)       1       0.5       Image: Codecs and the subscriber's ID         SIP Session Timer(seconds)       1       0.5       Image: Codecs and the subscriber's ID         SIP Session Timer(seconds)       1       0.5       Image: Codecs and the subscriber's ID         SIP Session Timer(seconds)       1       0.5       Image: Codecs and the subscriber's ID         SIP Session Timer(seconds)       1800       Image: Codecs and the subscriber's ID       Image: Codecs and the subscriber's ID         ID THF Payload(Scope:0-255)       101       Image: Codecs and the subscriber and the structure and the subscriber and the subscrif and the subscriber and the subscriber and the subscriber and th							which will be used for caller 1D display.
UDP Keep-alive Message       Enabled       ♥       ●       User Name         UDP Keep-alive Interva(seconds)       30       ●       ●       User account, provided by         Login Expire(seconds)       5600       ♥       ●       ●       ●         SIP Session Timer(seconds)       1       0.5       ●       ●       ●       ●         SIP Session Timer(seconds)       1       0.5       ●       ●       Notel server. If you light to Notel							SIP service subscriber's ID
UDP Keep-alve Message       Enabled       ✓       Vol Service provided by Vol Service provider.         UDP Keep-alve Interval(seconds)       30       ✓       Vol Service provider.         Login Expire(seconds)       3600       ✓       MAT Travesal         SiP Session Timer(seconds)       10.5       ✓       Prost         SiP Session Timer(seconds)       10.5       ✓       Notel server. If you login to Notel server, the value should be: com.notelnetworks.frewal         SiP Session Timer(seconds)       1800       ✓       Otometal server. If you login to Notel server, the value should be: com.notelnetworks.frewal         Subscribe Period(seconds)       1800       ✓       Otometal server. The value should be: com.notelnetworks.frewal         OTMF Type       RFC2833       ✓       ✓       Otometal server. The value should be: com.notelnetworks.frewal         DTMF Payload(Scope:96~255)       101       ✓       Otometal server. The value should be: com.notelnetworks.frewal         Subscribe Register       Dsabled       ✓       ✓       Otometal server. The value sends a subscribe packet to the server to subscribe hespice server subscribe hespice server subscribe hespice server subscribe hespice with server to subscribe hespice with server to subscribe hespice packet to the server to subscribe packet to the server subscribe pa	Advanced	>>					llcor Namo
Image: Code Repaired Intervalsections)       30       Image: Code Repaired Intervalsections)       30         Image: Code Repaired Intervalsections)       3600       2       Defines the STUN server will be active or not.         SUB       Subscribe Size Session Timer(seconds)       11       0.5       2         SIP Session Timer(seconds)       12       4       30       A special parameter just for Notel server. If you login to Notel server, if you login to		UDP Keep-alive Message	Enabled	~	0		User account, provided by
Login Explicite Control       3000       0       Defines the STUIN server will be active or not.         SiP Session Timer(seconds) T1       Disabled       0       Proxy Require         SiP Session Timer(seconds) T1       0.5       0       Nortel server. If you login to SiP Session Timer(seconds)         Mew to INFO DTMF       Disabled       0         DTMF Payload(Scope:96~255)       101       Afte Advanced parameters for administrator.         Subscribe Register       Disabled       0       Afte Advanced parameters for administrator.         Subscribe Register       Disabled       0       Deached       0         MWI Subscribtion Period(Scope:0~84600)       3600       The device sends a Subscribe Nesserve to subscribe Network Subscribe Serve to subscribe Networker Nesserve to subscribe Nereve		UDP Keep-alive Interval(seconds)	30				VoIP service provider.
S060       Image: S060       <		Login Expire(seconds)	3600		0		
SIP Session Timer(seconds) T1       0.5       Image: Control of the server of subscribe period (seconds) T2       Image: Control of the server of subscribe period (seconds) T2       Image: Control of the server of subscribe period (seconds) T4       Image: Control of the server of subscribe period (seconds) T4       Image: Control of the server of subscribe period (seconds) T4       Image: Control of the server of subscribe period (seconds) T4       Image: Control of the server of subscribe period (seconds) T4       Image: Control of the server of subscribe period (seconds) T4       Image: Control of the server of subscribe period (seconds) T4       Image: Control of the server of subscribe period (seconds) T4       Image: Control of the server of subscribe period (seconds) T4       Image: Control of the server of subscribe period (seconds) T4       Image: Control of the server of subscribe packet to the server of			5060		0		
SIP Session Timer(seconds)       11       U.S       Notel server. If you login to Notel server. If you login to Notel server, the value should be:         SIP Session Timer(seconds)       T4       5       Notel server. If you login to Notel server, the value should be:         SUBscribe Period(seconds)       1800       Image: Condex Server. If you login to Notel server, the value should be:       Notel server, the value should be:         SUBscribe Period(seconds)       1800       Image: Condex Server. If you login to Notel server, the value should be:       Notel server, the value should be:         Mutrice Period(seconds)       1800       Image: Condex Server. If you login to Notel server, the value should be:       Codes:         Mow to INFO DTMF       Disabled       Image: Condex Server. If you login to Uses.       Codes:       Choose the codecs you want to use.         Mow to INFO DTMF       Disabled       Image: Condex Server. If you login to Uses.       Advanced       The Advanced parameters for administrator.         100 reliable retransmission       Disabled       Image: Condex Server.       Image: Condex Server.       Advanced         Subscribe Register       Disabled       Image: Condex Server.       Image: Condex Server.       Image: Condex Server.         MVI Subscription Period(Scope:0~84600)       3600       Image: Condex Server.       Image: Condex Server.       Subscribe Packet to the server to subscribe Message waiting, the		RPort	Disabled	*	0		
be: com.nordenetworks.frewall be: com.nordenetworks.frewall Codecs com.nordenetworks.frewall Codecs Cod		SIP Session Timer(seconds) T1	0.5		0		Nortel server. If you login to
SIP Session Timer(seconds) T4       5       com.nortelhetworks.firewall         Subscribe Period(seconds)       1800       2       codecs         DTMF Type       RFC2833<		SIP Session Timer(seconds) T2	4				
DTMF Type RFC2833 Choose the codecs you want to use.   How to INFO DTMF Disabled Advanced   DTMF Payload(Scope:96~255) 101 The Advanced parameters for administrator.   100 reliable retransmission Disabled Q   Enable Precondition Disabled Q   Subscribe Register Disabled Q   Subscribe for MWI Disabled Q   MWI Subscription Period(Scope:0~84600) 3600 The device sends a Subscribe packet to the server to subscribe (will subscription Period(Scope:0~84600)   Caller ID Header FROM Q   Use Session Timer Disabled Q   Session Timer(seconds) Q		SIP Session Timer(seconds) T4	5				
DTMF Type       RFC2833 ▲       ✔       to use.         How to INFO DTMF       Disabled ▲       Advanced         DTMF Payload(Scope:96~255)       101       The Advanced parameters for administrator.         100 reliable retransmission       Disabled ▲       ✔         Enable Precondition       Disabled ▲       ✔         Subscribe Register       Disabled ▲       ✔         Subscribe for MWI       Disabled ▲       ✔         MWI Subscription Period(Scope:0~84600)       3600       The device sends a Subscribe packet to the server to subscribe market of the server to subscribe market of the server to subscribe market of the server after registration         Use Session Timer       Disabled ▲       ✔         Session Timer(seconds)       ✔       ✔		Subscribe Period(seconds)	1800		0		
DTMF Payload(Scope:96~255)       101       Advanced       The Advanced <t< th=""><th></th><th>DTMF Type</th><th>RFC2833</th><th>~</th><th>0</th><th></th><th></th></t<>		DTMF Type	RFC2833	~	0		
Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)       administrator.         Inter Payloal (Scope: 30x233)       Disabled       Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)         Inter Payloal (Scope: 30x233)       Disabled       Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)         Inter Payloal (Scope: 30x233)       Disabled       Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)         Inter Payloal (Scope: 30x233)       Disabled       Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)         Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)         Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)         Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)         Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)         Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)         Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)       Inter Payloal (Scope: 30x233)		How to INFO DTMF	Disabled	~			Advanced
100 reliable retransmission       Disabled       Image: Comparison of the comparison of t		DTMF Payload(Scope:96~255)	101				
Subscribe Register       Disabled       Image: Constraint of the server of the		100 reliable retransmission	Disabled	~	0		our minor a com
Subscribe for MWI       Disabled       Image: Constraint of the server to subscribe packet to the server to subscribe measure will send source will send source will send source will send source to the server to the server to the server to the server after registration         Caller ID Header       FROM       Image: Constraint of the device server to subscribe measure will send source to the server to the server to the server after registration         Use Session Timer       Disabled       Image: Constraint of the device server to the server after registration         Session Timer(seconds)       Image: Constraint of the device server to subscribe measure to the server after registration		Enable Precondition	Disabled	~	0		
MWI Subscription Period(Scope:0~84600)       3600       The device sends a Subscribe packet to the server to subscribe         Caller ID Header       FROM       ?         Use Session Timer       Disabled       ?         Session Timer(seconds)       ?		Subscribe Register	Disabled	~	0		
(seconds)     Soud     The device sends a Subscribe       Caller ID Header     FROM     ?       Use Session Timer     Disabled     ?       Session Timer(seconds)     ?		Subscribe for MWI	Disabled	~	0		
Caller ID Header     FROM     Image: College of the state of		MWI Subscription Period(Scope:0~84600) (seconds)	3600				
Session Timer(seconds)		Caller ID Header	FROM	*	0	Message waiting, the device	ce will
		Use Session Timer	Disabled	~	0	send a Subscribe packet t server after registration	to the
		Session Timer(seconds)			0		
Retresher Uac 🕑 🕜		Refresher	Uac	~	0		
Use user=phone Disabled 🕥 🕜		Use user=phone	Disabled	~	0		
Voice Encryption (SRTP) On 🕑 Off 🕜		Voice Encryption (SRTP)	🔘 On 🖲 Off	0			
ptime(ms) 20 💌 🕜		ptime(ms)	20	~	0		

To view the voicemail via the Phone interface:

1) Press Menu->Messages->Voice Mail->View Voice Mail.

1.Accour	nt1: Not	t registe	ered		
2,245:0	new O o	old Mail			
3.103:0	new O d	old Mail	U		
Back			Connect		

- 2) You can view the amount of the voice mail that includes new or old voice mail.
- 3) Choose the account and press the Connect button, then you are able to listen to your new and old messages.

To retrieve the new voicemail via the Phone interface:

- 1) Press the Voicemail hot key directly.
- 2) You may be prompted to enter the password which is needed to connect to your VoIP telephony system. It depends on your system.
- 3) Your voice mailbox is called and you are able to listen to your new and old messages.

#### Note:

Before retrieving the new voicemail, please make sure that the connecting code has been set on the phone.

### SMS

You can retrieve the SMS in the same way as Voicemail.

To retrieve the SMS via Phone interface:

- 1) Press Menu->Messages->Text Message to enter the configuration interface.
- 2) Use the navigation hot key to highlight the options. You can read the message in the Inbox/Sentbox/Outbox/Draftbox.

——————————————————————————————————————
1. New Message
2. Inbo×
3. Sentbox
Back Enter

3) Press View hot key to open and read the specific message.

To reply the SMS via Phone interface:

1) After retrieving the specific message, press Reply hot key, use the abc hot key to change the input method.

123	—From	n:251—	
Back			
	——To::	251——	
Back	abc	Del	Send

- 2) (Optional) If multiple lines are configured on your phone, select from which line messages are to be retrieved from the list using the navigation keys and Messages appear first to last in sent order.
- 3) Press Send hot key, you are required to choose an account which is used to send out the message and the number you want to send to.
- 4) Press Send hot key to send out the message, or Back hot key to return to the previous menu.

To edit a new message via the Phone interface:

1) Press Menu->Messages->Text Message->New Message to enter.

New Message						
Back	abc	Del	Send			

- 2) Edit the new message; use the abc hot key to change the Input Method.
- 3) (Optional) If multiple lines are configured on your phone, select from which line messages are to be retrieved from the list using the navigation keys and Messages

appear first to last in sent order.

4) Press Send hot key to send out the message, or Back hot key to return to the previous menu.

To delete the message via the Phone interface:

- 1) After retrieving the specific message, press Delete hot key.
- 2) You are prompted to confirm the Delete, press Yes hot key to delete the message, press No hot key to return to previous menu.



To edit the message via Web interface:

- 1) Choose Phone->SMS
- 2) Enter the account information, target phone number, message content.
- 3) Press Send button to send out the message, or Cancel button to cancel the operation.

#### Note:

You can not perform the messaging via Web interface except edit Message.

# **Using the Advanced Phone Functions**

# **Account Setting**

Please refer to the previous part "Configuration and Registration" for the basic Account setting information. The following table lists the instruction of the field about the advanced Account Setting.

Field Name	Description
UDP Keep-alive Message	Defines whether to active the phone UDP Keep-alive mechanism. The default is Enabled.
UDP Keep-alive Interval(seconds)	This parameter specifies how often the phone will send a packet to the SIP server. Default is 30 seconds.
<i>Login</i> <i>Expire</i> (seconds)	This parameter specifies the time frequency that phone refreshes its registration. The default interval is 3600 seconds.
Local SIP Port	Local SIP port. The default value is 5060.
Local RTP Port	Defines the local RTP port that the phone will listen and transmit. The default value is 11780.
RPort	The parameter allows you configuring the proxy to send responses back to a particular address and port. The default is disabled.
SIP Session Timer	This document defines an extension to the Session Initiation Protocol (SIP). This extension allows for a periodic refresh of SIP sessions through a re-INVITE or UPDATE request. The refresh allows both user agents and proxies to determine if the SIP session is still active.
<i>Subscribe</i> <i>Period</i> (seconds)	This parameter could set the period of the subscription. The default value is 3600.
DTMF Type	Select the DTMF type.

You can only configure these settings via Web interface.

- 1) Choose Account.
- 2) Select the desired account.
- 3) Choose Advanced interface to do the relating settings.

You can consult your system administrator for more information.

# TLS

TLS(Transport Layer Security), an IETF standards track protocol(RFC 5246), was based on the earlier SSL specifications developed by Netscape Corporation.

If you make a call based on TLS and SRTP (Set the transport as TLS and the SRTP as On), the IP phone UI will display the connecting and ring back interface. If the reception also set the transport as TLS, then IP Phone UI will display the icon  $\widehat{\mathbf{n}}$  on both side.

Go to Account->Basic, choose Transport option, in the pull-down menu, you can choose the TLS option, and then click the Confirm button to save the change.

	Status Account	Network	Phone	Contacts	Upgrad	le Security
Account	_	Account 1	~	_		NOTE
Basic >>						Display Name SIP service subscriber's na
DdSIC >>						which will be used for Call display.
	Register Status	Unknown	<b>C</b>			Register Name
	Account Active	C On	Off			SIP service subscriber's ID used for authentication.
	Label			0		User Name
	Display Name			0		User account, provided by VoIP service provider.
	Register Name			0		
	User Name			0		NAT Traversal Defines the STUN server
	Password			0		be active or not.
	SIP Server			Port 5060 ?		Proxy Require A special parameter just f
	Enable Outbound Proxy Server	Disabled	~	0		Nortel server. If you login
	Outbound Proxy Server			Port 5060 ?		Nortel server, the value si be:
	Transport	TLS	~	0		com.nortelnetworks.firew
	Backup Outbound Proxy Server				P,TCP,TLS three	
	NAT Traversal	Disabled	~	options. The i	registered packet OP, TCP or TLS, TL	co
	STUN Server			Port (Transport La	iyer Security)is	ranced
	Voice Mail					Advanced parameter administrator.
	Proxy Require			0		
	Anonymous Call	Off	*	0		
	On Code			0		
	Off Code			0		
	Anonymous Call Rejection	Off	*	0		

# **DNS-SRV**

If the SIP server cannot be used, the phone will be connected on the server which is available.

To set DNS-SRV via web interface:

Go to Account->Basic, choose Transport option, in the pull-down menu, you can choose the DNS-SRV option, and then click the Confirm button to save the change.

### **Using the Advanced Phone Functions**

alink _						Logo
	Status Account	Network	Phone	Contacts	Upgrade	Security
Account		Account 1	~		D NOTE	
						y Name rvice subscriber's nan
Basic >>						will be used for Calle
	Register Status	Unknown				
	Account Active	C On	Off		SIP se	ter Name rvice subscriber's ID
	Label			0	used f	or authentication.
	Display Name			0	User I	lame ccount, provided by
	Register Name			0		ervice provider.
	User Name			0		raversal
	Password			0		s the STUN server w ive or not.
	SIP Server			Port 5060 🕜	Proxy	Require
	Enable Outbound Proxy Server	Disabled	~	0	A spe	ial parameter just fo server. If you login t
	Outbound Proxy Server			Port 5060 🕜		server, the value she
	Transport	DNS-SRV	×	0		ortelnetworks.firewa
	Backup Outbound Proxy Server	UDP TCP	]	Port 5060 🕜	Code	
	NAT Traversal	TLS DNS-SRV		0	Choos to use	e the codecs you wa
	STUN Server	UNU UNU		Port 3478 🕜	Adva	red
	Voice Mail			0	The A	dvanced parameters
	Proxy Require			0	admin	strator.
	Anonymous Call	Off	~	0		
	On Code			0		
	Off Code			0		
	Anonymous Call Rejection	Off	~	0		

# **Network Setting**

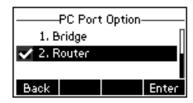
# **PC Port Setting**

Please refer to the previous part "Configuration and Registration" for the basic Network WAN setting information. The following table lists the instructions of the field about the Network PC Port Setting.

Field Name	Description
As Bridge	If you select the Bridge mode, then the two Fast Ethernet ports will be transparent.
As Router	If you select the Router mode, the SIP phone will work as a router.
IP address	User could configure the PC port IP address.
DHCP Server	If you set the DHCP server on, the device connected to the PC port will get the IP address automatically between the start IP address and the end IP address. But if you select the bridge mode, the DHCP server can not work.
Start IP Address	Indicate the range of the IP address.
End IP Address	Indicate the range of the IP address.

To configure PC Port settings via Phone interface:

- 1) Press Menu->Settings->Advanced Settings.
- 2) Enter the password required, scroll to Network option, then press Enter hot key, select PC port and press Enter hot key.



- 3) If you choose Bridge, it will return to the previous menu.
- 4) If you choose Router, you will be prompted to enter the IP Address, Subnet Mask, DHCP Server Disable/Enable.
- 5) Press Save hot key to save the changes, or Back hot key to return to the previous menu.

To configure PC Port settings via Web interface:

Choose Network-> PC Port to do the relating configuration. You can consult your system administrator for more information.

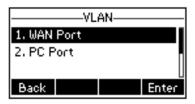
# **VLAN Setting**

VLAN is a group of hosts with a common set of requirements that communicate as if they were attached to the Broadcast domain, regardless of their physical location. The following table lists the instruction of the field about the VLAN Setting.

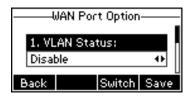
Field Name	Description
QoS	When the network capacity is insufficient, QoS could provide priority to users by setting the value.
Local RTP Port	Define the port for voice transmission.
WebServer	Users can choose the WebServer type: Disable, HTTP, HTTPS, or HTTPS & HTTP.

To configure VLAN settings via Phone interface:

- 1) Press Menu->Settings->Advanced Settings.
- 2) Enter the password required, scroll to Network option, then press Enter hot key, select VLAN and press Enter hot key.



3) Choose WAN Port, press Enter hot key.



- 4) Set the VLAN Status, and input the VID Number, Priority.
- 5) Press Save hot key to save the settings, or Back hot key to return to VLAN menu.
- 6) Follow the same way to set the PC Port option.

To configure VLAN settings via Web interface:

Choose Network->Advanced to do the relating configuration. You can consult your system administrator for more information.

### LLDP

The Link Layer Discovery Protocol (LLDP) is a vendor-neutral Layer 2 protocol that allows a network device to advertise its identity and capabilities on the local network.

Enable LLDP function; the phone will go to switch to get related VLAN parameters automatically. (Synchronous with VALAN in switch)

To configure LLDP settings via Web interface:

- 1) Choose Network->Advanced->LLDP->Active option, in the pull-down menu, choose enable.
- 2) Then enter the corresponding Packet Interval in Packet Interval field.
- 3) You can also disable this function when you choose disable in active field.
- 4) Click the Confirm button to save the change.

Enterprise IP Phone	e		<u>Using th</u>	<u>e Advanc</u>	ed Phon	e Function
Yealink						<u>Logout</u>
Status	Account	Network	Phone	Contacts	Upgrade	Security
	Ir	iternet Port (WAN)	PC Port Ad	vanced		
LLDP 🕜 VLAN 🥝 WAN Port	Active Packet Interval Active VID	Disabled 120 Disabled 0	<ul> <li>(Scope:1-</li> <li>(0-4094)</li> </ul>	~3600s)	area ne that ex single t group o given s	I is a logical local twork (or LAN) tends beyond a raditional LAN to a of LAN segments, pecific rations.
PC Port	USRPRIORITY Active VID USRPRIORITY	0 Disabled 0	(0-4094)		capacit QoS co priority	the network y is insufficient, uld provide to users by the value.
Port Link	WAN Port Link PC Port Link	auto neg auto neg				<b>XTP Port</b> the port for voice ssion.
VPN 🕜	VPN Active Upload VPN Config	Disabled Imp		浏览		
Voice QoS	Voice QoS	40	(0~63)			

### HTTPS

This IP phone can support HTTPS (Hypertext Transfer Protocol over Secure Socket Layer). Adding SSL layer under HTTP, in short, it is a security version of HTTP. Users can set this transmission mode via web page.

To configure HTTPS settings via Phone interface:

- 1) Press Menu->Settings->Advanced Settings.
- 2) Enter the password required, scroll to Network option, press Enter hot key, select Webserver Type option, then press Enter hot key again.

———Webserver Type———						
1. Webserver Type: HTTP & HTTPS						
Back		Switch	Save			

- 3) Press the navigation keys or Switch hot key to choose the transmission mode.
- 4) Press Save hot key to save the settings, or Back hot key to return to Webserver Type menu.

To configure HTTPS settings via Web interface:

Go to Network->Advanced, choose WebServer option, in the pull-down menu of Type field, choose the transmission mode, and then click the Confirm button to save the changes.

### **Using the Advanced Phone Functions**

, a to a hts,
pice
c

#### Note:

- 1. For more details of the HTTPS, you can consult with your system administrator.
- 2. IP phone also support Internet Protocol Version 6.

# **Maintenance Tasks**

Enterprise IP Phone

### **Administrator Mode**

The phone allows two modes to configure the phone:

- User Mode
- Administrator Mode

Administrator mode grants unlimited access to the phone configuration on both Web and Phone interface. User Mode is not able to access the settings on the Phone interface such as: Accounts, Network, Reset to Factory, other advance phone settings.

### Administrator/User Password

Administrator mode grants unlimited access to the phone configuration on both web and phone user interface. The administrator/user password is used to access:

- Web interface.
- the advanced settings of the phone such as Network, Account, Reset to Factory

Settings via the Phone interface.

The default administrator password is **admin**. Meanwhile the user name for Web interface access is **admin**.

To change the administrator password via phone interface:

- 1) Press the Menu ->Settings-> Advanced Settings.
- 2) You are prompted to enter the required password, the default one is **admin**.
- 3) Scroll to Set Password, press Enter hot key.
- 4) You are prompted to enter the Current, New and Confirm password, Press abc hot key to change the input method.
- 5) Press Save hot key to confirm the change, or Back hot key to return to the previous menu.

To change the administrator password via web interface:

Choose Security->Password->admin, enter the Current, New and Confirm password, click Confirm button to save the changes, or Cancel button to cancel the changes.

To logout via Web interface:

Click the Logout button in the top right corner.

# Reboot

You should reboot the phone when you are challenged, e.g. after applying changes to the phone configuration.

To reboot via Web interface:

- 1) Choose Upgrade->Basic.
- 2) Click Reboot button.
- 3) You are prompted to confirm the change, press OK to confirm the changes, press Cancel to cancel the operation.

#### Note:

Please do not power off during reboot, or it will cause the flash memory error.

# **Reset to Factory**

You should reset the phone only in this case: the phone configuration was changed and the phone is not functioning anymore. To maintain the configuration of the phone, you need your system administrator or service provider's advice.

To reset to factory via phone interface:

- 1) Press the Menu->Settings->Advanced Settings
- 2) You are prompted to enter the required password, the default one is **admin**.
- 3) Scroll to Reset to factory option, and press Enter hot key to enter the configuration interface.

### **Using the Advanced Phone Functions**

4) You are prompted to confirm the change, press OK hot key to reset to factory settings, or press Cancel hot key to return to previous menu.

——Advar	nced Setti	ng-	
Reset To P	Factory S	ett	ing?
Cancel			Ok

5) It will take a few minutes to reset, please do not power off during resetting, or it will cause flash memory error.

#### Note:

The reset to factory option must in the admin mode.

To reset to factory via Web interface:

- 1) Choose Upgrade->Basic.
- 2) Click Reset button.
- 3) You are prompted to confirm the change, press OK to confirm the changes, press Cancel to cancel the operation.

#### Note:

If you confirm all current setting changes including contact list, call history, account settings, etc will be lost, you need to export the configuration first if you still want to import the old configurations after reset. Or your phone must be configured a new manually unless mass provisioning is used!

To Export/Import the old configuration file via Web interface:

- 1) Choose Upgrade->Advanced, select Import/Export Configuration, choose Export button to export the file to your local computer.
- Choose Upgrade->Advanced, select Import/Export Configuration, choose Browse button, select the specific configuration file in your local computer, choose Import button.
- 3) It will take a few minutes to reset, please do not power off during resetting, or it will cause flash memory error.

# Firmware Update

The phone is delivered with pre-installed firmware which allows operating your phone flawlessly. If you require updating the phone's firmware please contact your system administrator. You can only update the firmware via Web interface.

#### Warning:

1. Please do not power off or unplug the Ethernet cable during the updating.

To update the firmware manually via Web interface:

#### **Using the Advanced Phone Functions**

- 1) Choose Upgrade->Basic->Browse, select the firmware file in your local computer.
- 2) Click Upgrade button to update the new firmware.

To update the firmware automatically via Web interface:

- Choose Upgrade->Advanced, configure the relating settings: Custom Option, Custom Option Type, URL, Account, Password, Common AES Key and MAC-Oriented AES Key, PNP config and Check New Config etc.
- 2) Choose Confirm button, the phone will check the server for a new firmware in a specific time, and it updates automatically if there is new firmware.
- 3) You can also update the firmware immediately by pressing Auto-provision button.

Set Auto Provision via phone interface:

- 1) Go to Menu->Settings->Advanced Settings-> Auto Provision.
- 2) Enter the URL, User Name and Password.
- 3) Click the Save hot key to save the changes.

#### Note:

- 1. Any power interruption during the following process will most likely lead to a flash memory error. As a result the system cannot boot up anymore..
- 2. The upgrade priority is PNP, Custom Option, URL by descending.
- 3. Users can also delete some configuration options by Auto-provision, for example, to delete the admin password.

# Decryption

This IP phone can support y00000000005.cfg and mac.cfg files encryption and decryption for user authentication to realize security usage. If there are any encrypted y00000000005.cfg or mac.cfg files on the server, users can open the webpage of your IP phone. Go to Upgrade->Advanced, choose and fill in the Common AES Key (for y0000000005.cfg, should be 16 bit) and MAC-Oriented AES Key (for mac.cfg, should be 16 bit) option, then click the Confirm button to decryption the files and upgrade to the new version. Shown as below:

### **Using the Advanced Phone Functions**

Yealink						-	<u>Loqout</u>
	Status	Account	Network	Phone	Contacts	Upgrade	Security
			Basic	Advanced			
	Custom C URL Account Password Common MAC-Orie ForbidZer WaitTime PNP Conf Check Ne Click this immediat	AES Key nted AES Key o a ig w Config button to auto prov	String String Enable 5 Enable Disable	d v d v d v	<b>?</b> 浏览?	Spe you pro Pro abo AE3 It is Clitic pro Clitic pro Con Con con the all t Sys The exp	TE stom Option wight the DHCP Option that want to use for visioning. Refer to Auto vision Manual for details ut provisioning. S Key provided by ISP. k this button to auto vision immediately. vision immediately. button fles to backup settings, and could import he settings after reset. stem Log re are two methods to ort the system log, Local server.
	Export Sy	/stem Log	Local	Export			
	PCAP Tra	Confirm	Star	t Stop	Export		

#### Note:

You can ask your system administrator for the decrypt password.

Set AES Key via phone interface:

1) Go to Menu->Settings->Advanced Settings->Set AES Key.



- 2) Enter the Common AES and the MAC-oriented option.
- 3) Press the Save hot key to save the changes.

## **Zero-sp-touch**

Zero-sp-touch this function can help users to configure AUTOP and network parameters quickly.

Enable this function, when the power is on or press the corresponding DSSKEY, the phone will turn to the zero-sp-touch interface.

Turn on Zero-sp-touch via web interface:

1) Choose Upgrade->Advanced->Zero Active, in the pull-down menu, choose enable to turn on this function.

#### 2) Click Confirm to save.

Yealink					-	<u>Loqout</u>
	atus Account	Network	Phone	Contacts	Upgrade	Security
		Basic	Advanced			
	Custom Option(128 ~ 254) Custom Option Type URL Account Password Common AES Key MAC-Oriented AES Key ForbidZero WaitTime PNP Config Check New Config Click this button to auto pro immediately Export / Import Config	String String Enabled Disabled vision Auto	d v Q d v Q o provision Q	0	Spr you Pro abbo AE It i It Cli pro Co Co Co Co Co Co Co Co Co The all ti Sy Th Sy Th exp	ATE stom Option acify the DHCP Option that a want to use for visioning. Refer to Auto visioning. Refer to Auto vision immediately source of the store this button to auto vision immediately. button files to backup riguration files to backup ri
	Export System Log	Local	× Q			
	PCAP Trace	Start	t Stop	Export		

Enter into zero-sp-touch interface, first a countdown interface come into view,

- 1) Not any operation or press cancel hot key, will enter idle interface.
- 2) Press status key enters into phone's network interface, can search the internet.
- 3) Press OK key, enter a network setting interface, press next key enter an AutoP setting interface, enter the corresponding contents; press OK key to save the settings. Press back key return to previous menu.

# System Log Export

If there are any errors happened in your phone, you can export the system log and send to your system administrator for diagnosis.

To export the System Log:

- Choose Upgrade->Advanced, select Export System Log type, if the type is Local, it will export the syslog directly; if the type is server, it will export the syslog to the specified server.
- 2) Click Export button to export the file

# **PCAP Trace Export**

The PCAP Trace used to record the date transport of your IP phone. If there are any errors happened in your phone, you can export the PCAP trace and send to your system

administrator for diagnosis.

To export the PCAP Trace:

Choose Upgrade->Advanced to enter, select PCAP Trace option, click Start button began to capture the trace, and click Stop to stop capture the trace, and then click Export to export the file to your local computer.

# 802.1X

IEEE 802.1X is an IEEE Standard for port-based Network Access Control (PNAC). It is part of the IEEE 802.1 group of networking protocols. It provides an authentication mechanism to devices wishing to attach to a LAN, either establishing a point-to-point connection or preventing it if authentication fails. It is used for securing wireless 802.1X access points and is based on the Extensible Authentication Protocol (EAP).

This IP phone can support 802.1X. For the details, please consult your system administrator.

# **DSS keys Configuration**

The phone has 3 line keys which are able to set up to 35 functions per key. The following list shows the functions you can set on the DSS keys and provides a description for each function. The default configuration for each key is Line.

- N/A
- Conference
- Forward
- Transfer
- Hold
- DND
- Redial
- Call Return
- SMS
- Pick Up
- Call Park
- DTMF
- Voice Mail
- Speed Dial
- Intercom
- Line
- BLF
- URL
- Group Listening
- Public Hold
- Private Hold

**Using the Advanced Phone Functions** 

- Shared Line
- XML Group
- Group Pickup
- Paging
- Record
- Xml Browser
- Hot Desking
- URL Record
- LDAP
- BLF List
- Prefix
- Zero-sp-touch
- ACD
- Local Group
- Broadsoft Group

#### Note:

 Quick access features like Intercom and Voicemail must first be configured on your PBX in order to work on your phone. See your system administrator for more information.

#### Conference

You are allowed to configure the DSS key to be used as a conference key while remaining in the current call. This key allows a user on a call to conference another party while remaining in the conference.

To assign the key as Conference:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Conference from the list.
- 2) Press Confirm button to save the changes.

## Forward

If the key is configured as Forward key, press this key under the idle status, the IP phone will turn to the forward page, and you can set the Forward to number, then when there is any call to the extension number will be forwarded to the set number automatically.

To assign the key as Forward:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Forward from the list.
- 2) Enter the extension number you want to forward to in the Extension field.

**Using the Advanced Phone Functions** 

3) Press Confirm button to save the changes.

## Transfer

You are able to configure the key as a transfer key to perform the Blind/Attended/Semi-Attended Transfer.

To assign the key as Transfer:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, and choose Transfer from the list.
- 2) Enter the Number in the "Value" field, when you are on a conversation, press this key, the phone will Blind transfer to the number. Or you can leave it black to set as the transfer button.
- 3) Press Confirm button to save the changes.

# Hold

The key can be configured as a hold key. You can use this key to hold and retrieve a call during the conversation.

To assign the key as Hold:

- 6) Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Hold from the list.
- 7) Press Confirm button to save the changes.

# DND

If the key is configured as DND key, you are allowed to active the DND function immediately when you press it. Press it again to deactivate DND mode.

To assign the key as DND:

- 1) Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose DND from the list.
- 2) Press Confirm button to save the changes.

# Redial

If the key is configured as Redial key, press this key under the idle status, it will enter the Dialed Calls interface, then you can choose a special line to call out by pressing the line keys.

To assign the key as Redial:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Redial from the list.
- 2) Press Confirm button to save the changes.

# **Call Return**

When the key is configured as Call Return key you are allowed to dial out the last phone call you received.

To assign the key as Call Return:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Call Return from the list.
- 2) Press Confirm button to save the changes.

# SMS

When the key is configured as SMS key you are allowed to access SMS quickly by pressing this key.

To assign the key as SMS:

- 1) Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose SMS from the list.
- 2) Press Confirm button to save the changes.

# Pick Up

When you configure a Pick Up key, you specify the extension that you want to monitor. Then, when the monitored extension receives a call, you can press this key to pick up the incoming calls.

To assign the key as Pick Up:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Pick Up option from the list.
- Enter the feature code (for example, input \*78345, \*78 is the feature code and the 345 is the extension number you want to pickup) in the Value field.
- 3) In the "Line" field, select a line for which to apply this key.
- 4) Press Confirm button to save the changes.

# Call Park

Call Park is a feature that allows a person to put a call on hold at one telephone set and continue the conversation from any other telephone set.

The "call park" feature is activated by pressing a preprogrammed button or a special sequence of buttons. When the conversation which is monitored was transferred to an unused extension number, you can press this key to retrieve the call.

To assign the key as Call Park:

### **Using the Advanced Phone Functions**

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Call Park from the list.
- 2) Enter the number you want to park in the Value field.
- 3) In the "Line" field, select a line for which to apply this key.
- 4) Press Confirm button to save the changes.

# DTMF

You are allowed to send out the desired DTMF number during the conversation. The number needs to be set in advance.

To assign the key as DTMF:

- 1) Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose DTMF from the list.
- 2) In the "Value" field, enter the specific number.
- 3) Press Confirm button to save the changes.

# Voice Mail

When the key is configured as Voicemail key you are allowed to access voicemail quickly by pressing this key.

To assign the key as Voice Mail:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Voice Mail from the list.
- 2) Enter the number you want to set as the voice mail box in the Value field.
- 3) In the "Line" field, select a line for which to apply this key.
- 4) Press Confirm button to save the changes.

# **Speed Dial**

You can configure the key as a simplified speed dial key. This key function allows you to easily access the most frequently dialed numbers.

To assign the key as Speed Dial:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Speed Dial from the list.
- 2) Enter the number you want to perform Speed Dial in the Value field.
- 3) In the "Line" field, select a line for which to apply this key.
- 4) Press Confirm button to save the changes.

# Intercom

You can configure the key for Intercom mode and is useful in an office environment as a quick access to connect to the operator or the secretary.

To assign the key as Intercom:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Intercom from the list.
- 2) Enter the extension number you want to intercom in the Value field.
- 3) In the "Line" field, select a line for which to apply this key, the default one is Line 1.
- 4) Press Confirm button to save the changes.

## Note:

Your VoIP PBX must support this feature. And make sure the Intercom Allow is enable.

# Line

You can set these keys as line keys to active up to the three user accounts.

To assign the key as Line:

Choose Phone->DSS Key-> Line Key, choose one of the link key you want to make the assignment, there is a pull-down menu in the Type field, choose Line from the list, press Confirm button to save the changes.

# BLF

You can configure the key for Busy Lamp Field (BLF) which allows you to monitor the status (idle, ringing, or busy) of other SIP accounts. User can dial out on a BLF configured key.

To assign the key as BLF:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, and choose BLF from the list.
- 2) Enter the number you want to monitor in the Value field,
- 3) In the "Line" field, select a line for which to apply this key.
- 4) And then enter the feature codes in the extension field.
- 5) Press Confirm button to save the changes.

Please refer to "LED Instruction" for more details about the LED status in different situation.

## Note:

In the Web interface, you can also set the pickup number to active the pickup function. For example, if you set the BLF number as 212, and the pickup number is \*83, then

**Using the Advanced Phone Functions** 

when there is an incoming call to 212, press the BLF key, it will call out the \*83 automatically to pickup the incoming call on 212.

# URL

If the key is configured as URL, then pressing this key, you can send HTTP requests to a web server.

To assign the key as URL:

- Choose Phone->DSS Key->Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, and choose URL from the list.
- 2) Enter the number you want to set as URL in the Value field.
- 3) Press Confirm button to save the changes.

# **Group Listening**

When the key is configured as Group Listening key, you are allowed to enable the Speakerphone and Handset/Headset mode at the same time. It is suitable for the group conversation which has more than one person at one side. You are able to speak and listen using handset/headset; meanwhile the others nearby can listen using speakerphone. You can get back to the previous mode by pressing the key again. (If the current mode is handset or headset, users can press the speaker button to open or close the group listening function)

To assign the key as Group Listening:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, and choose Group Listening from the list.
- 2) Press Confirm button to save the changes.

# **Public Hold**

The key can be configured as a public hold key. During a conversation, all members belonging to that particular BLA group can use this key to hold or retrieve a call.

To assign the key as Public Hold:

- Choose Phone->DSS Key->Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Public Hold from the list.
- 2) Press Confirm button to save the changes.

# **Private Hold**

The key can be configured as a private hold key. During a conversation, all members

belonging to that particular BLA group can use this key to hold the call, but only the initiator can retrieve the call.

To assign the key as Private Hold:

- 1) Choose Phone->DSS Key->Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Private Hold from the list.
- 2) Press Confirm button to save the changes.

# **Shared Line**

The Shared Line Appearances (SLA, which is also named as BLA) feature allows subscribers to share SIP lines and also provides status monitoring of the shared line. When a user places an outgoing call using such an appearance, all members belonging to that particular SLA group are notified of this usage and are blocked from using this line appearance until the line goes back to idle state or when the call is placed on hold. Similarly all members of the SLA group are notified of an incoming call and the call can be picked up on a line appearance associated with the SLA extension.

To assign the key as Shared Line:

- 1) Choose Phone->DSS Key->Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Shared Line from the list.
- 2) Enter the condition code in the Value field.
- 3) In the "Line" field, select a line for which to apply this key, the default one is Line 1.
- 4) Press Confirm button to save the changes.

# XML Group

The keys can be configured as XML Group key. Then pressing this key under the idle status, you can enter the XML Group interface.

To assign the key as XML Group:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose XML Group from the list.
- 2) In the Line field, choose a remote phone book to specify the record, so you must configure the remote phone book at first.
- 3) Press Confirm button to save the changes.

# **Group Pick up**

When you configure a Group Pick Up key, you specify the extension group that you want to monitor. Then, when the monitored group receives a call, you can press this key to pick up the incoming call. If the group receives multiple calls simultaneously, you will pick up the specific one the server assigns to you.

To assign the key as Group Pick Up:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, and choose Group Pick Up option from the list.
- 2) Enter the feature code (for example,\*78) in the Value field.
- 3) In the "Line" field, select a line for which to apply this key.
- 4) Press Confirm button to save the changes.

# Paging

You can configure the key as Paging key. When you press this key, the phone will dial the number out directly.

To assign the key as Paging:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Paging option from the list.
- 2) Enter the number you want to dial out directly in the Value field.
- 3) In the "Line" field, select a line for which to apply this key.
- 4) Press Confirm button to save the changes.

# Record

Call recording is a phone function to record the conversation in the process of dialogue. Using this feature, please pay attention to the maximum recording time and frequency in advance. Generally, it maybe a few minutes.

To assign the key as Record:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Record option from the list.
- 2) Press Confirm button to save the changes.

# **Xml Browser**

Xml browser is a simple browser function, which is based on xml language and http/https service, users can use tools such as php, javascript, etc., accordance with the established syntax, to generate server-side functions dynamically to meet the needs of the user xml file, and then downloaded it to sip phone by http/https, so as to realize a simple browser features. Using xml browser, customers can achieve the personalized features, such as weather information, stock information, date of inquiry, access to address book, google search, news browsing, playing music, configuration of the terminal parameters and other functions.

To assign the key as Xml Browser:

1) Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Xml Browser from

the list.

- 2) Enter the URL address you want to connect to in the Value field.
- 3) Press Confirm button to save the changes.

# **Hot Desking**

Hot desking 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 at all, which means actual personal offices would be often vacant, consuming valuable space and resources.

Hot desking is a phone feature that allows accounts to login or logout in an IP phone. After a certain account landing, the corresponding configuration of the account will be applied to the phone.

This IP phone can support two kinds of Hot desking: Base Mode and Advanced Mode.

- Base Mode: Pressing the DSS key which is set to be Hot Desking when the phone is idle to active the Base Mode. By this mode, it will clear all of the account records, then set up an account and register for the first account.
- Advanced Mode: By this mode, it will clear the first account information, then connect the URL to download the Xml file and configure a new account.

To assign the key as Hot Desking:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Hot Desking option from the list.
- 2) In the "Value" field, enter the URL address in the input field to active the advanced mode. Or you can leave it black to active the base mode.
- 3) Press Confirm button to save the changes.

# **URL Record**

During the conversation, pressing the type of DSS key, and then follow the voice prompts to achieve the call recording capability.

- 1) When you are on the conversation, pressing the DSS key to start the recording process in the current Call.
- 2) Enable the recording function, the recording icon will be flashing for the ongoing recording process.
- 3) Pressing the DSS key again to disable the recording function, and the flashing recording icon will be disappeared simultaneously.
- 4) Follow the voice prompts to listen to the recording.

To assign a DSS key as URL Recorder:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose URL Recorder from the list.
- 4) Enter the condition code in the Value field.
- 5) Press Confirm button to save the changes.

#### Note:

During a conversation, press this type of DSS key to start the recording process; if the other party hung up, your phone will turn to the idle status.

# LDAP

If you want to use the LDAP function, you must configure the corresponding options in the LDAP interface at first.

To assign the key as LDAP:

- 1) Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose LDAP from the list.
- 2) Press Confirm button to save the changes.
- 3) When you press this key under the idle, the phone will turn to the contact search interface.

# **BLF** List

BLF list is a function which can monitor the group status, it is not one to one monitoring, but the information feedback from the sever to decide which BLF list will monitor which account.

To set BLF List via web interface:

- 1) Choose Account->Advanced-> BLF List URI, enter the BLF List URI.
- 2) Then enter the BLF List Code in the BLF List Code field.
- 3) Click the Confirm button to save.

To assign the key as BLF List:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, and choose BLF List from the list.
- 2) In the "Line" field, select a line for which to apply this key.
- 3) Press Confirm button to save the changes.

# Prefix

When you set up the function of prefix, press this key, the phone will be ready to make a new call, and show up the content which your set previously on the dail interface. And you could enter other figure and call out

To assign the key as Prefix:

 Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, and choose Prefix from the list, and enter the number you want to show up on the dial interface in the Value field.

2) Press Confirm button to save the changes

## Zero-sp-touch

You can also press the DSSKey which set as the Zero-sp-touch. Then press the DSSKey, the phone will turn to the Zero-sp-touch interface.

To assign the key as Zero-sp-touch:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, and choose Zero-sp-touch from the list.
- 2) Press Confirm button to save the changes

# ACD

ACD(Automatic Call Distribution) is automatic call distribution equipment, is according specific Transfer Rules and distribution strategy to switch the access call to the right person.

Presses DSSKey, it will pop up a login box, enter the User ID and Password, click the Log In hot key.

Presses DSSKEY again enter to the ACD Status page, choose Available/Unavail to change the status.

You can also press LogOut hot key to logout.

When you status is Available, the calls will be directed to your phone. Or the status is unavailable, the calls will not be directed to your phone.

To assign the key as ACD:

- Choose Phone->DSS Key-> Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, and choose ACD from the list.
- 2) In the "Line" field, select a line for which to apply this key.
- 3) Press Confirm button to save the changes

#### Note:

ACD is not available on all call servers. For more information, contact your system administrator.

# Local Group

The keys can be configured as Local Group key. Then pressing this key under the idle status, you can enter the Local Group interface.

To assign the key as Local Group:

- 1) Choose Phone->DSS Key->Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Local Group from the list.
- 2) In the Line field, choose a group.
- 3) Press Confirm button to save the changes.

# **Broadsoft Group**

The keys can be configured as Broadsoft Group. Then pressing this key under the idle status, you can enter the Broadsoft Group interface.

To assign the key as Broadsoft Group:

- Choose Phone->DSS Key->Line Key, choose one of the keys you want to make the assignment, there is a pull-down menu in the Type field, choose Broadsoft Group from the list.
- 2) In the Line field, choose a broadsoft Group to specify the record, so you must configure the Broadsoft Group at first.
- 3) Press Confirm button to save the changes.

# **Tone Settings**

You can define the frequency and time period of all the following tones:

- Dial
- Ring Back
- Busy
- Congestion
- Call Waiting
- Dial Recall
- Record
- Info
- Stutter
- Message
- Auto Answer

To edit the tone filed via Web interface:

- 1) Choose Phone->Tones.
- 2) Enter the frequency and time period(in ms) as the following format: Frequency /Time Period (for example 400/200).
- 3) Press Confirm button to save the changes, Cancel to cancel the change.

## Note:

1. Please contact your system administrator for more information about the

**Using the Advanced Phone Functions** 

frequency and time period parameters. You can enter up to 8 groups for each tone.

2. If the frequency is set as 0, it means silence.

# Voice

To edit the Voice filed via Web interface:

- 1) Choose Phone->Voice.
- 2) Set the following parameters shown in the table.

Parameter	Description
Echo canceller	Defines whether to enable the echo canceller.
VAD	Voice activity detection (VAD), also known as speech activity detection or speech detection, is a technique used in speech processing in which the presence or absence of human speech is detected.
CNG	A comfort noise generator (CNG) is a program used to generate background noise for voice communications during periods of silence that occur during the course of conversation.
JITTER BUFFER	It is a shared data area where voice packets can be collected, stored, and sent to the voice processor in evenly.
Туре	To choose the type of JITTER BUFFER, adaptive or Fixed.
Delay	To set the Min Delay, Max Delay and Normal Delay parameter.

Yealink							LOQU
Easy vop	Status	Account	Network	Phone	Contacts	Upgrade	Security
Pre	eference   Featu	res Softkey Layo	ut DSS Key	Action URL Voi	ice Ring To	ones Dial Plan	SMS
	Cancellation 《	Echo canceller VAD CNG Type Min Delay Max Delay Normal Delay	Enable Disable Enable C Ada 0 300	d 💌 🕜		CNG Comfor Genera JITTER It is a s where be colle sent to	
		Confirm	[	Cancel			

3) Press Confirm button to save the changes, Cancel to cancel the changes.

# Ring

Users can group your contacts, and then set the ringing tone for each group. To edit the Ring option via Web interface:

1) Choose Phone->Ring.

Yealink	
Ease vor Status Account Network Phone Contacts Upgrade	Security
Preference   Features   Softkey Layout   DSS Key   Action URL   Voice   Ring   Tones   Dial Plan   SM	ims
1 Internal Ringer Text	
Internal Ringer File Ring1.wav	
2 Internal Ringer Text	
Internal Ringer File Ring1.wav 💌	
3 Internal Ringer Text	
Internal Ringer File Ring1.wav	
4 Internal Ringer Text	
Internal Ringer File Ring1.wav	
5 Internal Ringer Text	
Internal Ringer File Ring1.wav	
6 Internal Ringer Text	
Internal Ringer File Ring1.wav	
7 Internal Ringer Text	
Internal Ringer File Ring1.wav 💌	
8 Internal Ringer Text	
Internal Ringer File Ring1.wav 💌	
9 Internal Ringer Text	
Internal Ringer File Ring1.wav	
10 Internal Ringer Text	
Internal Ringer File Ring1.wav	
Confirm	

- 2) Internal Ringer Text: To set group name. For example, family.
- 3) Internal Ringer File: To choose a special ring tone for the group.
- 4) Click the Confirm button to save the changes.

# **Trouble Shooting**

#### I can not register to the server?

- 1) Check the IP address. If you set your WAN port in DHCP mode, please make sure that your DHCP server is on.
- 2) Check your gateway.
- 3) Check your DNS server.
- 4) Make sure your account information is the same as you have got from your ISP.
- 5) Check whether the SIP server is on.
- 6) Check the SIP register port, the default value is 5060.

#### I can't get the IP address?

- 1) Make sure you have plugged the Ethernet cable into the WAN port.
- 2) Make sure that the DHCP server is on, and there are available IP addresses in the server.
- 3) Try to set your WAN port to static IP client mode.

#### During a call, I can not hear any voice?

- 1) Make sure your handset is tightly connected with the phone.
- 2) Check whether you have muted the conversation or not.
- 3) Consult the outbound server details with your ISP.

#### Have DTMF problem?

- 1) Check which kind of DTMF you are using, and whether it is compatible with the server.
- 2) Consult the payload value with your ISP.

#### How to change the time?

Select the time zone or enter the time information manually on the webpage or the phone.

#### How to answer the incoming calls during a call?

If a call comes in when you are in a conversation, press the Answer hot key to answer the call, or press the Reject hot key to refuse it.

#### How to refuse incoming calls during a call?

You can turn off the function of call waiting, and then our phone will refuse all the

incoming calls when you are in a conversation.

#### How to send SMS?

You could edit the SMS in the Menu-> Messages->Text Messages.

Note:

Make sure that the SIP server you have registered supports SMS function.

#### How to update the firmware?

- 1) Enter the webpage of your phone, go to Upgrade, then you can find the option" Select and Upgrade Firmware" at the bottom of the page.
- 2) Select the file to update, then click the Upgrade button.

#### Note:

Make sure the firmware you choose is provided by your service provider, or the device will probably crash after the update.

#### How to auto provision?

Consult the auto provision server address with your ISP.

The manual is only for reference; please take the object as the standard.

We reserve the right to improve or change the product and the user guide without notice.

You can download the latest user manuals from our website: <u>http://www.yealink.com/en/download.asp?BigClassName=IP%20Phone</u>