

Enterprise IP Phone

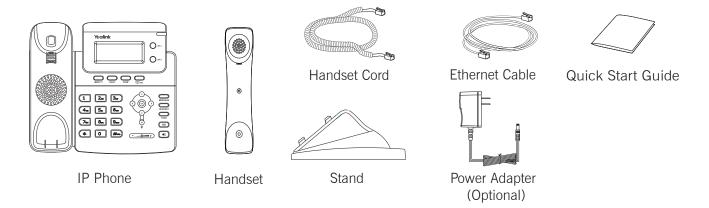
SIP-T20P & SIP-T20



Quick Start Guide (V73.40)

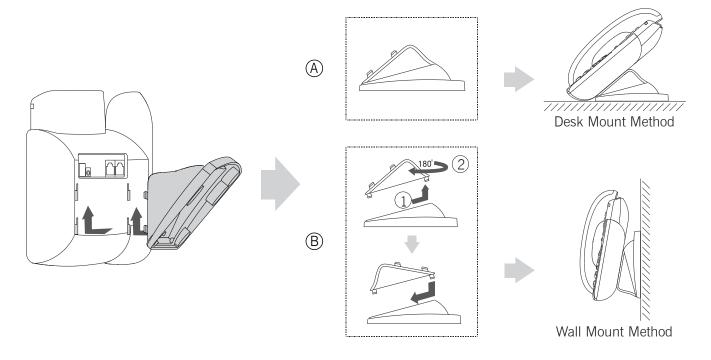
Packaging Contents

The following items are included in your package. If you find anything missing, contact your system administrator.

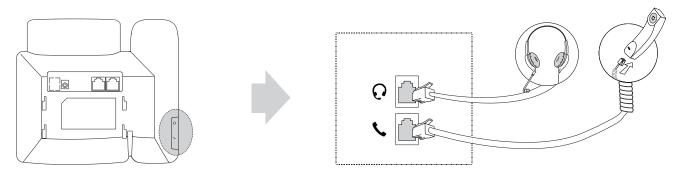


Assembling the Phone

1. Attach the stand, as shown below:



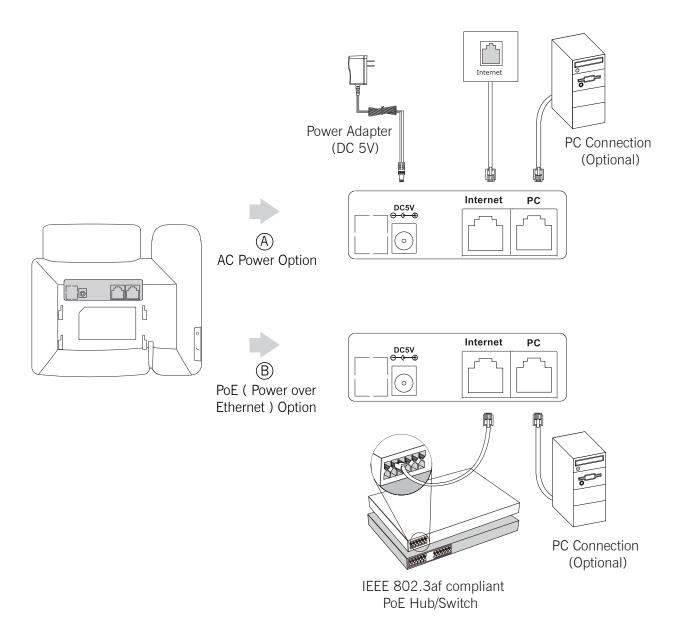
2. Connect the handset and optional headset, as shown below:



Note: The headset should be purchased separately.

3. Connect the network and power, as shown below:

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



Note: PoE is only applicable to the SIP-T2OP IP phone. If inline power (PoE) is provided, you do not need to connect the power adapter. Make sure the hub/switch is PoE-compliant.

Startup

After the IP phone is connected to the network and supplied with power, it automatically begins the initialization process. After startup, the phone is ready for use. You can configure the phone via web user interface or phone user interface.

Initializing
Please wait

Configuring Your Phone

Configuring via web user interface

Accessing the web user interface:

- 1. Press the **OK** key when the phone is idle to obtain the IP address of the phone.
- 2. Open a web browser on your computer, enter the IP address into the address bar (e.g., "http://192.168.0.10" or "192.168.0.10") and press **Enter**.
- 3. Enter the user name (default: admin) and password (default: admin) in the login page and click Confirm.

Network Settings: Click on Network-> Basic-> IPv4 Config

You can configure the network in the following ways:

DHCP: By default, the phone attempts to contact a DHCP server in your network to obtain its valid network settings, e.g., IP address, subnet mask, gateway address and DNS address.

Static IP Address: If your phone cannot contact a DHCP server for any reason, you need to configure IP address, subnet mask, gateway address, primary DNS address and secondary DNS address for the phone manually.

PPPoE: If you are using an xDSL modem, you can connect your phone to the Internet via PPPoE mode. Contact the ITSP for the PPPoE user name and password.

Note: The IP phone also supports IPv6, but IPv6 is disabled by default.

Wrong network settings may result in inaccessibility of your phone and may have an impact on the network performance. Contact your system administrator for more information.

Account Settings: Click on Account-> Register-> Account X (X=1,2)

Parameters of the account:

Register Status: It shows the register status of the current account.

Line Active: You can select Enabled/Disabled to enable/disable the account.

Label: It is shown on the LCD screen to identify the account.

Display Name: It is shown as caller ID when placing a call.
User Name: It is provided by ITSP for registration (required).

Register Name: It is an authenticated ID for authentication provided by ITSP (required).

Password: It is provided by ITSP for registration (required). Server Host: It is provided by ITSP for registration (required).

Note: Check with your system administrator if any error appears during the registration process or a specific configuration is required for your registration.

Configuring via phone user interface

Network Settings:

Press the MENU key when the phone is idle, select **Settings->Advanced Settings** (default password:admin)->**Network->WAN Port /VLAN/Webserver Type/802.1x Settings/VPN/LLDP** to configure the network.

Account Settings:

Press the **MENU** key when the phone is idle, select **Settings->Advanced Settings** (default password:admin)->**Accounts** to configure the account.

Note: For more information on account parameters, refer to Configuring via web user interface.

Basic Call Features

Placing a Call

Using the handset:

- 1. Pick up the handset.
- 2. Enter the number, and then press #sen

Using the speakerphone:

- 1. With the handset on-hook, press ()
- 2. Enter the number, and then press #seno)

Using the headset:

- 1. With the headset connected, press \bigcirc to activate the headset mode.
- 2. Enter the number, and then press $\boxed{\textbf{\#}_{\text{\tiny{SEND}}}}$.

Note: During a call, you can alternate between headset, hands-free speakerphone and handset modes by pressing the **HEADSET** key, the **Speakerphone** key or by picking up the handset. Headset mode requires a headset connected.

Answering a Call

Using the handset:

Pick up the handset.

Using the speakerphone:

Press ().

Using the headset:

Press —.

Note: You can reject an incoming call by pressing the X key.

Ending a Call

Using the handset:

Hang up the handset or press MENUA.

Using the speakerphone:

Press or MENU?

Using the headset:

Press MENU 2.

Redial

- Press RD to enter the placed calls list, press or to select the desired entry, and then press RD or #BD .
- Press RD twice when the phone is idle to dial the last dialed number.

Call Mute and Un-mute

- Press (x) to mute the microphone during a call.
- Press (x) again to un-mute the call.

Call Hold and Resume

To place a call on hold:

Press during an active call.

To resume the call, do one of the following:

• If there is only one call on hold, press

• If there is more than one call on hold, press • or • to select the desired call, and then press • or • to select the desired call, and then press • or • to select the desired call, and then press • or • to select the desired call, and then press • or • to select the desired call, and then press • or •
Call Transfer You can transfer a call in the following ways:
Blind Transfer
1. Press during a call. The call is placed on hold.
2. Enter the number you want to transfer to.
3. Press .
Semi-Attended Transfer
1. Press during a call. The call is placed on hold.
2. Enter the number you want to transfer to, and then press #
3. Press when you hear the ring-back tone.
Attended Transfer
1. Press during a call. The call is placed on hold.
2. Enter the number you want to transfer to, and then press #
3. Press when the second party answers.
Call Forward
To enable call forward:
1. Press when the phone is idle, and then select Features->Call Forward.
2. Select the desired forward type:
Always Forward Incoming calls are forwarded unconditionally. Busy Forward Incoming calls are forwarded if the phone is busy. No Answer Forward Incoming calls are forwarded if not answered after a period time.
3. Enter the number you want to forward to. For No Answer Forward , press or to select the desired ring time to wait before forwarding from the After Ring Time field.
4. Press ok to accept the change.
Call Conference
1. Press during an active call. The call is placed on hold.
2. Enter the number of the second party, and then press $\#_{\text{sec}}$.
3. Press again when the second party answers. All parties are now joined in the conference.
4. Press the Cancel soft key to disconnect all parties.
Speed Dial
To configure a speed dial key:
1. Press when the phone is idle, and then select Features-> DSS Keys .
2. Select the desired line key, and then press $(o\kappa)$.
3. Select Speed Dial from the Type field, select the desired line from the Account ID field, and enter the number in the
Value field.
4. Press (ok) to accept the change.
To use the speed dial key: Press the speed dial key to dial out the preset number.

Voice Message Message waiting indicator on the idle screen indicates that one or more voice messages are waiting at the message center. The MESSAGE key LED lights up. To listen to voice messages: 1. Press or ok 2. Follow the voice prompts to listen to your voice messages. **Customizing Your Phone Call History** 1. Press (▲) when the phone is idle, press (▲) or (▼) to scroll through the list. 2. Select an entry from the list, you can do the following: Press #send to call the entry. \bullet Press $\overset{\longleftarrow}{\ \ \ \ \ }$ to add the entry to the local directory. • Press (×) to delete the entry from the list. **Contact Directory** To add a contact: 1. Press when the phone is idle, and then select Local Directory->All Contact. 2. Press (◆) or (▶) to select **Enter** and then press (oĸ). 3. Press ♠ or ♥ to select New Item and then press (oĸ). 4. Enter a unique contact name in the Name field, and enter the phone number in the proper field.

About us

Founded in 2001, Yealink, the global TOP 3 SIP phone provider and a leading provider of VoIP Phone and IP communication solutions, has been focusing on VoIP products characterized by reliable quality, easy deployment and affordable price for more than a decade. Today, customers from over 140 countries rely on Yealink as the backbone of global collaboration to extend the value of network communications.

More about Yealink

Since 2001, Yealink has continued to develop state-of-the art, hi-tech VoIP communication terminals that include IP video phones, IP phones and USB phones. With a worldwide market in more than 140 countries, the company has also established a first-class international service network.

Our priorities are quality, functionality, ease-of-use, customer support and competitive pricing. To provide new solutions for the future in this fast-evolving sector, Yealink's large, talented and highly-experienced VoIP R&D team is totally committed to the pursuit of excellence.

To date, we have passed more than 80 interoperability tests and meet certification requirements set by leading VoIP system and platforms. Yealink works in close partnership with T-Mobile, Telefonica, Portugal Telecom and other leading telecommunication service providers.

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