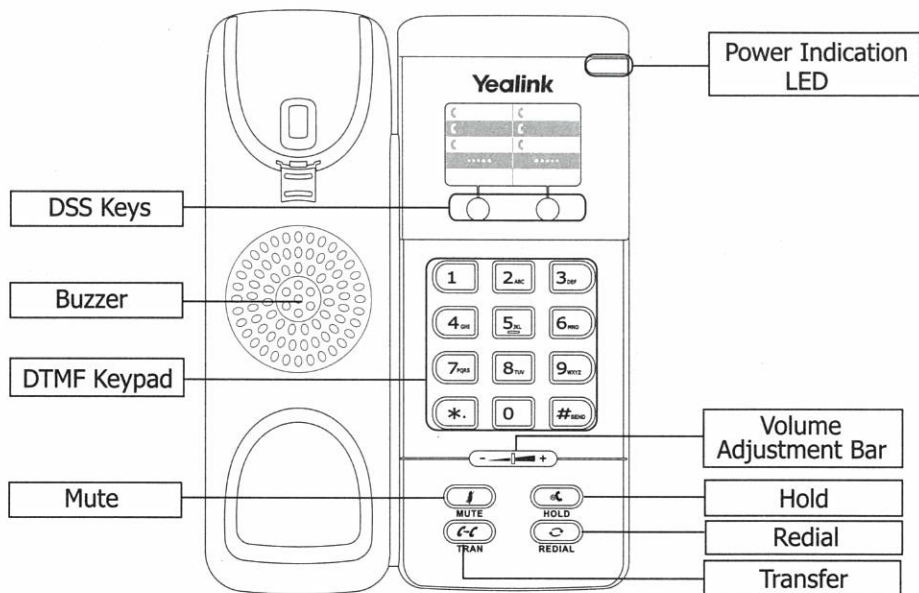


Simply IP Phone SIP-T18P



Quick Reference

www.yealink.com



Power Indication LED

Table 1 Green Light

LED Status	Description
Steady Green	The phone is under the idle status
Slow Blinking	There is call on mute
Fast Blinking twice	The phone is under DND/Forward status

Table 2 Red Light

LED Status	Description
Steady Red	There is new voicemail to your extension
Slow Blinking	There is an incoming call to the device
Fast Blinking twice	There is call on hold

Table 3 Orange Light

LED Status	Description
Steady Orange	Power on
Blinking Orange	Network unavailable, the account is not registered or registered failed
Steady Orange	Updating

Making a Call

Pick up the handset to make a call.

- **Dial number directly:** Dial the number directly, and press the # key to call out
- **Re-dial:** Dial the RD button to dial out the last called number directly

Answering a Call

Answering an incoming call in the following ways:

- If you are on another call, press the Hold key to answer it
- If you are not on another call, just lift the handset to answer it
- You can also press the MUTE button to deny the incoming call

Muting a Call

- Press the MUTE button during a conversation to mute the call, and press it again to get the microphone return to normal conversation

Volume Adjustment

- During the call, you can press the volume adjustment bars to adjust the volume

Putting a Call on Hold

- Press the HOLD button to put your active call on hold
- If there is only one call on hold, press the HOLD key again to retrieve it
- If there are two calls on hold, press HOLD to put the current call on hold, and retrieve the third party

Voice Mail

- The presence of new Voice Mail messages will be indicated by the LED, users can pick up the handset and dial the concatenated code to get the voicemail in the server

Conference Call

- Press the Hold key during an active call
- The first call is placed on hold. You can dial the third number, then press # key to call out.
- When the call is answered, press the Redial key to establish a three-party conference.

Hang up to disconnect the conference call

Call Transfer

There are three ways to transfer the call: Blind Transfer, Attended Transfer and Semi-Attend Transfer.

Blind Transfer:

- A and B is on an conversation, A press TRAN button to put B on hold, then A can dial the third number C and press TRAN button to call out. Then A will turn to idle status
- After C answered it, B can talk to C

Attended Transfer:

- A and B is on an conversation, A press TRAN Button to put B on hold, then A can dial the third number C and press the # button to call out
- After C answered it, A and C can have a private conversation without B hearing it, then A hung up or press the TRAN button to complete the transfer
- A will be disconnected from the call. B can talk to C

Semi-Atten Transfer:

- A and B is on an conversation, A press TRAN button to put B on hold, then A can dial the third number C and press the # button to call out
- While C is ringing, A hang up or press the TRAN button to complete the transfer.
- You will be disconnected from the call, when C pick up, B can talk to C

Check the IP Status

- IP Address: Pick up the handset and dial "***90#", follow the voice prompt to get the IP address
- Subnet Mask: Pick up the handset and dial "***91#", follow the voice prompt to get the Subnet Mask
- Gateway: Pick up the handset and dial "***92#", follow the voice prompt to get the Gateway
- DNS1: Pick up the handset and dial "***93#", follow the voice prompt to get the DNS1
- DNS2: Pick up the handset and dial "***94#", follow the voice prompt to get the DNS2

Note:
The functions such as DSS key, Call Forward, DND, Call Waiting, etc. should be set via web page.
For the details, please refer to the user manual.

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.

V0.2