



Product Name: Yealink T19P E2 IP Phone (SIP-T19P)

Manufacturer: -

Model Number: SIP-T19P

Please Note: The Yealink SIP-T19P has been discontinued. Please see T30 Range as an

alternative.

Yealink T19P E2 IP Phone (SIP-T19P)

Please Note: Yealink T19P IP phones are PoE (Powered-Over-Ethernet) and do not include a mains Power Supply. If mains power is required, this can be added via the drop down menu above.

The Yealink SIP-T19P E2 Single Line VoIP Phone is an entry-level VoIP phone featuring a 132 x 64-pixel graphical LCD, dual 10/100 Ethernet ports and full-duplex speakerphone. Yealink T19P (SIP-T19P) Key Features

� 1 Line Appearances

� Dual-port 10/100 Ethernet

ï¿⅓ Single SIP account

ï¿1/2 Full-duplex hands-free speakerphone

� No expansion modules supported

آزرًا RJ9 Headset PortDesktop with footstand (detachable), Wall-mountable

ï¿⅓ Supports PoE, Power Supply available separately

Dual 10/100 Mbps network ports with integrated PoE are ideal for extended network use. The SIP-T19P supports a single VoIP account, simple, flexible and secure installation options, plus IPv6 and SRTP/ HTTPS/ TLS, VLAN and QoS. It includes headset use, is wall-mountable and has been designed very specifically for better business.

Yealink T19P IP Phone (SIP-T19P) - Technical Specifications Audio Features

� Codec: Opus, G.722, G.711(A/µ), G.729AB, G.726, iLBC

าั¿½ DTMF: In-band, Out-of-band (RFC 2833) and SIP INFO

าั¿1/2 Full-duplex hands-free speakerphone with AEC

ï¿1/2 VAD, CNG, AEC, PLC, AJB, AGC

Phone Features

� 1 VoIP account

� Call hold, mute, DND

� One-touch speed dial, hotline

ï¿1/2 Call forward, call waiting, call transfer

ï¿⅓ Group listening, SMS

� Redial, call return, auto answer

ï¿1/2 Local 3-way conferencing

� Direct IP call without SIP proxy

� Ring tone selection/import/delete

12.1/2 Set date time manually or automatically

� Dial plan

ï¿1/2 XML Browser, action URL/URI

ï¿1/2 Integrated screenshots



- � RTCP-XR
- � Enhanced DSS Key

Directory

- � Local phonebook up to 1000 entries
- تز½ Blacklist
- � XML remote phonebook
- ï¿⅓ Smart dialing
- � Phonebook search/import/export
- า๊¿1/2 Call history: dialed/received/missed/forwarded

IP-PBX Features

- � Anonymous call, anonymous call rejection
- � Hot-desking, emergency call
- � Message Waiting Indicator (MWI)
- � Voice mail, call park, call pickup
- ï¿1/2 Intercom, paging, music on hold
- � Call completion
- ï¿⅓ Flexible seating
- � Executive and Assistant
- ï¿1/2 Centralised call recording
- ï¿⅓ Visual Voice Mail

Display and Indicator

- ï¿1/2 132x64-pixel graphical LCD
- تز½ LED for call and message waiting indication
- � Intuitive user interface with icons and soft keys
- � Multilingual user interface
- � Caller ID with name and number
- � Power saving

Feature Keys

- � 6 features keys: message, headset, redial, mute, hands-free speakerphone
- ï¿1/2 5 navigation keys
- ï¿1/2 Volume control keys

Interface

- � 2 x RJ45 10/100M Ethernet ports
- าั¿½ Power over Ethernet (IEEE 802.3af), class 1
- � 1 x RJ9 (4P4C) handset port
- � 1 x RJ9 (4P4C) headset port

Other Physical Features

- ï¿1/2 Wall mountable
- ī¿½ External Yealink AC adapter (optional):AC 100~240V input and DC 5V/600mA output
- � Power consumption (PSU): 0.9-1.3W
- � Power consumption (PoE): 1.2-1.8W
- � Dimension (W*D*H*T):185mm*188mm*143mm*38mm



- � Operating humidity: 10~95%
- � Operating temperature: -10~50°C (+14~122°F)

Management

- าั¿½ Configuration: browser/phone/auto-provision
- � Auto provision via FTP/TFTP/HTTP/HTTPS for mass deploy
- � Auto-provision with PnP
- � Zero-sp-touch,TR-069
- � Phone lock for personal privacy protection
- � Reset to factory, reboot
- � Package tracing export, system log

Network and Security

- ï¿1/2 SIP v1 (RFC2543), v2 (RFC3261)
- � Call server redundancy supported
- � NAT traversal: STUN mode
- تز½ Proxy mode and peer-to-peer SIP link mode
- � IP assignment: static/DHCP
- � HTTP/HTTPS web server
- � Time and date synchronisation using SNTP
- ï¿1/2 UDP/TCP/DNS-SRV (RFC 3263)
- آذِيً QoS: 802.1p/Q tagging (VLAN), Layer 3 ToS DSCP
- ï¿1/2 SRTP for voice
- � Transport Layer Security (TLS)
- ï¿1/2 HTTPS certificate manager
- ï¿1/2 AES encryption for configuration file
- � Digest authentication using MD5/MD5-sess
- ï¿⅓ IEEE802.1X
- ۱₽v6 ½ Pv6
- � LLDP/CDP/DHCP VLAN
- ICE ½'زï

Package Features

آذِ½ Package content:- Yealink SIP-T19P E2 IP phone- Handset with handset cord- Ethernet Cable (1.5m CAT5E UTP Cable)- Stand- Quick Start Guide- Power Adapter (Optional)

Please Enquire

Options available for Yealink T19P E2 IP Phone (SIP-T19P):

Power Supply Required?

None - Using PoE, UK PSU required (+£7.10).

RPS Extension ID

TEXT.

RPS Extension Password

TEXT.



RPS Display Name

TEXT.

RPS PBX URL

TEXT.

RPS NAT Enabled

TEXT.

RPS Outbound Proxy

TEXT.

RPS SIP Port

TEXT.

RPS Enable Outbound Proxy

TEXT.

RPS Own Server URL

TEXT.

RPS Own Server Name

TEXT.

Need Product Setup / Training / Support?

Not Required.

Compatible Headsets

Yealink YHS33 IP Headset (+£35.10), JPLConnect-2 binaural headset and disconnect cord (+£76.25), Not Required, JPL Connect-1 monaural headset and disconnect cord (+£65.20).