

Yealink T20P IP Phone (SIP-T20P)



Product Name: Yealink T20P IP Phone (SIP-T20P)

Manufacturer: -

Model Number: SIP-T20P

Yealink T20P IP Phone (SIP-T20P)

Please Note: The Yealink SIP-T20P has been discontinued. For an alternative, we recommend the Yealink T21P IP Phone (SIP-T21P).

The Yealink T20P is an entry level IP Phone featuring a 2x15 character LCD screen, 2 VoIP accounts, 31 keys and boasting 3-way conference ability. Yealink T20P Key Features

- 2-line display.
- 5 hard keys for controlling the various functions.
- Simple & convenient user interface
- Reliable
- Excellent price-performance ratio.

The T20P is an easy phone to install and inexpensive to set up. It can be reliably integrated into a variety of SIP-based systems. It is well suited to environments where basic business phone features are required.

- Supports 2 VoIP lines, each with a separate configuration.
- Delivers superior voice clarity, supports advanced wideband voice codec G.722 and incorporates full-duplex hands-free speakerphone with enhanced AEC.
- The T20P provides 2 programmable keys, 23 hard keys and 2 indication LEDs.
- Supports SMS, online advertisement, weather report and groupable enterprise phone book.

Note: Please refer to the Yealink VoIP Phone Range Detailed Comparison Table for more information.

Yealink T20P Technical Specifications
Phone Features

- 2 VoIP accounts, Hotline, Emergency call
- Call hold, Call waiting, Call forward, Call return
- Call transfer (blind/semi-attended/attended)
- Caller ID display, Redial, Mute, DND
- Auto-answer, three-way conferencing
- Speed dial, voicemail
- Tone scheme, volume control
- Programmable soft keys
- Message Waiting Indication (MWI) LED
- Direct IP call without SIP proxy
- Ring tone selection/import/delete
- 300 Entry phonebook, blacklist
- Call history: dialed/received/missed/forwarded
- Menu-driven UI
- Localised language and input method

VoIP PBX System Integration

- Busy lamp field (BLF), BLF list
- Bridged line appearance (BLA)

Yealink T20P IP Phone (SIP-T20P)

- DND and Forward synchronisation
- Call park, call pickup
- Intercom, paging, music on hold
- Call recording, call completion
- Group listening, group pickup
- Anonymous call, anonymous call rejection
- Network conference
- Distinctive ringtone
- Dial Plan, dial-now

Voice Features & Codecs

- Full-duplex speakerphone with AEC
- Wideband codec: G.722
- Narrowband codec: G.711µ/A, G.723.1
- G.726, G.729AB
- VAD, CNG, AEC, PLC, AJB, AGC

Network Features

- SIP v1 (RFC2543) & v2 (RFC3261)
- IPV6
- DNS SRV (RFC3263)
- NAT Traversal: STUN mode
- Redundant server support
- DTMF: In-Band, RFC2833, SIP Info
- Proxy mode & peer-to-peer SIP link mode
- IP Assignment: Static/DHCP/PPPoE
- TFTP/DHCP/PPPoE client
- Bridge/router mode for PC port
- Telnet/HTTP/HTTPS server
- NAT/DHCP server
- DNS client

Phone Management

- Auto-provision via FTP/TFTP/HTTP/HTTPS
- Auto-provision with PnP
- TR069 optional
- SNMP V1/2 optional
- Configuration of browser/phone/auto-provision
- Customised factory configuration
- Trace package and system log export
- Action URL & active URI

Phone Connectivity

Physical Specification

- TI TITAN chipset
- 2 x 15 characters lines
- 3-line LCD with an icon line and
- 31 keys (including 9 function keys)
- 4 LEDs: 1 x power, 2 x line, 1 x message

Yealink T20P IP Phone (SIP-T20P)

- 2 x RJ45 10/100M Ethernet ports
- 1 x RJ9 handset port
- 1 x RJ9 headset port
- Power adapter: AC 100~240V input and DC 5V/1.2A Output (Optional)
- Power consumption: 1.4-2.6W
- Power over Ethernet (IEEE 802.3af)
- Net weight: 0.77kg
- Dimension: 185 x 200 x 90mm
- Operating humidity: 10~95%
- Storage temperature: up to 60°C
- Wall-mountable

Yealink T20P Standards Certifications

Please Enquire

Options available for Yealink T20P IP Phone (SIP-T20P) :

Power Supply Required?

[EU PSU required \(+£3.99\)](#), [None - Using PoE](#), [UK PSU required \(+£3.99\)](#).

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

RPS Own Server URL

RPS Own Server Name