Enterprise IP Phone   SIP-T19P

Entry-level IP Phone with 1 Line

- 132 x 64-pixel graphical LCD
- Single VoIP account
- Two-port 10/100 Ethernet Switch, integrated PoE
- Full-duplex speakerphone
- Headset, wall mountable
- XML remote phonebook

The SIP-T19P is one of Yealink's latest answers for the entry-level IP phone that offers features and performance normally associated with much more advanced phones. The quite intentional choice of high-quality materials, combined with a generously large 132 x 64-pixel graphical LCD that gives a clear 5-line display, guarantees both a smoother user experience and easy access to much more visual information at a glance. 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
EASY VoIP

Audio Features

> Full-duplex hands-free speakerphone with AEC
> Codecs: G.711(A/μ), G.723, G.729AB, G.726
> DTMF: In-band, out-of-band(RFC 2833) and
   SIP INFO
> VAD, CNG, AEC, PLC, AJB, AGC

Directory

> Local phonebook up to 1000 entries
> Black list
> XML remote phonebook
> Intelligent search method
> Phonebook search/import/export
> Call history: dialed/received/missed/forwarded

Phone Features

> 1 VoIP account
> Call hold, mute, DND,
> One-touch speed dial, hotline
> Redial, call return, auto answer
> Call forward, call waiting, call transfer
> Group listening, SMS
> Local 3-way conferencing
> Direct IP call without SIP proxy
> Ring tone selection/import/delete
> Keypad lock, emergency call
> Set date time manually or automatically
> Dial Plan, XML Browser, action URL&action URI

IP-PBX Features

> Anonymous call, anonymous call rejection
> Message waiting indicator (MWI)
> Voice mail, call park, call pickup
> Intercom, paging, music on hold
> Call completion, hot-desking

Display and Indicator

> 132x64-pixel graphical LCD
> LED for call and message waiting indication
> Intuitive user interface with icons and soft keys
> National language selection
> Caller ID with name, number

Network and Security

> SIP v1 (RFC2543), v2 (RFC3261)
> IPv6
> NAT transverse: STUN mode
> Proxy mode and peer-to-peer SIP link mode
> IP assignment: static/DHCP/PPPoE
> HTTP/HTTPS web server
> Time and date synchronization using SNTP
> UDP/TCP/DNS-SRV(RFC 3263)
> QoS: 802.1p/Q tagging (VLAN), Layer 3 ToS, DSCP
> SRTP for voice
> Transport Layer Security (TLS)
> HTTPS certificate manager
> AES encryption for configuration file
> Digest authentication using MD5/MD5-sess
> IEEE802.1X

Management

> Configuration: browser/phone/auto-provision
> Auto provision via FTP/TFTP/HTTP/HTTPS
   for mass deploy
> Auto-provision with PnP
> BroadSoft device management
> Zero-sp-touch
> Provisioning server redundancy supported
> Reset to factory, reboot
> Package tracing export, system log

Physical Features

> 2xRJ45 10/100M Ethernet ports
> 29 keys including 4 soft keys
> 1xRJ9 handset port
> 1xRJ9 headset port
> Wall mountable
> Power adapter: AC 100~240V input and
> DC 5V/600mA Output
> Power over Ethernet (IEEE 802.3af)
> Power consumption（PSU）: 1.24-2.62W
> Dimension (W*D*H*T):
   185MM*188MM*143MM*38MM
> Operating humidity: 10~95%
> Operating temperature: -10~50°C

Package Features

> Qty/CTN: 10 PCS
> N.W/CTN:
> G.W/CTN:
> Giftbox size: 215MM*200MM*121MM
> Carton Meas: 630MM*436MM*210MM

Certifications

CE FC RoHS REACH ISO 9001

3CX Asterisk AVAYA BROADSOFT Technology Partner

Genesys