

Executive IP Phone with 6 Lines & HD Voice

- TI TITAN chipset and TI voice engine
- HD Voice: HD Codec, HD speaker, HD handset
- 320x160-pixel graphical LCD
- Two-port 10/100 Ethernet Switch, integrated PoE
- 6 VoIP accounts
- Headset(wired/wireless), Expansion module



Yealink SIP-T28P represents the next generation VoIP phone which designed for the business user who needs rich telephony features, friendly UI and super voice quality. It is equipped with the TI TITAN chipset, offers high definition voice quality through TI voice engine, HD handset, HD speaker and HD codec (G.722). By the large, high-resolution graphical display, and together with all the 48 keys, SIP-T28P offers an excellent user experience to configure, make calls, express XML browser, etc. Moreover, to avoid problems with unwanted violations of your audio data, Yealink SIP-T28P supports the security standards TLS, SRTP, HTTPS, 802.1x, Open VPN and AES encryption which are necessary to protect against electronic eavesdropping and data theft.

Audio Features

- > Full-duplex hands-free speakerphone with AEC
- > Codecs: G.722(Wideband), 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/LDAP remote phonebook
- > Intelligent search method
- > Phonebook import/export
- > Call history: dialed/received/missed/forwarded

Phone Features

- > 6 VoIP accounts
- > 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
- > Soft keys programmable

Interface

- > 2xRJ45 10/100M Ethernet ports
- > Power over Ethernet (IEEE 802.3af) , class 2
- > 1xRJ9 (4P4C) handset port
- > 1xRJ9 (4P4C) headset port
- > 1XRJ12 (6P6C) EXT port:
Supports up to 6 Expansion Modules for an attendant console application

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, call recording, hot-desking

Display and Indicator

- > 320x160-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, open VPN
- > Call server redundancy supported
- > 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
- > Zero-sp-touch, SNMP, TR069
- > Reset to factory, reboot
- > Package tracing export, system log

Other Physical Features

- > TI TITAN chipset
- > 48 keys including 16 freely programmable keys
- > 4 context-sensitive "soft" keys
- > Power adapter: AC 100~240V input and DC 5V/1.2A Output
- > Power consumption: 1.3-3.1W
- > Net weight: 1.01kg
- > Dimension(W*D*H*T):
273mm*200mm*158mm*58.5mm
- > Operating humidity: 10~95%
- > Storage temperature: -10~50°C

Package Features

- > Qty/CTN: 5 PCS
- > N.W/CTN: 6.9KG
- > G.W/CTN: 7.6KG
- > Giftbox size: 285mm*240mm*110mm
- > Carton Meas: 580mm*295mm*250mm

Certifications

