Enterprise HD IP Phone   SIP-T22P

Professional IP phone with 3 Lines & HD voice

- TI TITAN chipset and TI voice engine
- 3 VoIP accounts, 132x64 graphic LCD
- HD Voice: HD Codec, HD speaker, HD handset
- IPV6, BLF/BLA, XML Browser, Hot-desking
- 2xLAN, PoE, Headset, Wall-mountable

Yealink SIP-T22P features intuitive user interface and enhanced functionality which make it easy for people to interact and maximize productivity. With TI TITAN chipset and TI leading VoIP voice engine, it enables enhanced high-definition audio, outsourced management options, flexible deployment and third-party communications applications. As a cost effective IP solution, it helps users to streamline business processes, delivery a powerful, security and consistent communication experience for small and large offices environment.

Yealink
easy VoIP

Enterprise HD IP Phone   SIP-T22P

Phone Features

3 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, 3-way conferencing
Speed dial, SMS, Voicemail
Message Waiting Indication (MWI) LED
Tone scheme, Volume control
Direct IP call without SIP proxy
Ring tone selection/import/delete
Phonebook (300 entries), Black list
Call history: dialed/received/missed/forwarded
Menu-driven user interface
Localized language and input method
Soft keys programmable

Advanced Features

XML phonebook search/import
LDAP phonebook
XML Browser
PUSH XML
Action URL & Active URI
Headset（Call center mode）
Multicast IP Paging

IP PBX System Integration

Busy lamp field (BLF), BLF list
Bridged line appearance (BLA)
DND&Forward synchronization
Intercom, Paging, Music on hold
Call park, Call pickup
Call recording, Call completion
Group listening, Group pickup
Anonymous call, Anonymous call rejection
Network conference
Distinctive ringtone
Dial Plan, Dial-now

Codecs and Voice Features

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

Network Features

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

Management

Auto-provision via FTP/TFTP/HTTP/HTTPS
Auto-provision with PnP
SNMP V1/2 , TR069 optional
Configuration: browser/phone/auto-provision
Factory configuration customized
Trace package and system log export

Security

802.1x, VLAN QoS (802.1pq)
Transport Layer Security (TLS)
HTTPS (server/client), SRTP (RFC3711)
Digest authentication using MD5/MD5-sess
Secure configuration file via AES encryption
Phone lock for personal privacy protection
Admin/VAR/User 3-level configuration mode

Physical Features

TI TITAN chipset
132x64 graphic LCD
32 keys including 4 soft keys
5 LEDs: 1xpower, 3xline, 1xmessage
1xRJ9 handset port
1xRJ9 headset port
2xRJ45 10/100M Ethernet ports
Wall-mountable
Power adapter: AC 100~240V input and
DC 5V/1.2A Output
Power over Ethernet (IEEE 802.3af)
Power consumption: 1.4-2.6W
Net weight: 0.77kg
Dimension: 185 x 200 x 90mm
Operating humidity: 10~95%
Storage temperature: up to 60°C

Package Features

Qty/CTN: 10 PCS
N.W/CTN: 12.020KG
G.W/CTN: 14.100KG
Measurement: 0.083CMB
Carton Meas: 565x550x235MM

Certifications

RoHS Compliant
CE
FC
ISO 9001
REACH
MIC
BROADSOFT Technology Partner
Asterisk
GENESYS AN ALCATEL·LUCENT COMPANY
AVAYA DevCONNECT TESTED

DC5V
Internet
PC
POWER
Internet

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EASY VoIP

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