Gigabit Color IP Phone   SIP-T32G

Professional Gigabit phone with Color LCD

- Dual-port Gigabit Ethernet, Power over Ethernet
- 3" TFT-LCD, 262K colors, Intuitive user interface
- TI Aries chipset and TI voice engine
- HD Voice: HD Codec, HD speaker, HD handset
- 3 VoIP accounts, BLF/BLA, IPv6, Open VPN
- Headset, Wall mountable

SIP-T32G IP Phone is Yealink latest innovation for managers with demanding collaborative communication needs. For managers, it is a future-proofing network investment which supports seamless migration to GigE-based network infrastructure. Dual-port Gigabit Ethernet is designed for flexible deployment options and lower cabling expenses. With its high-resolution TFT color display, SIP-T32G IP Phone offers a brilliant depiction of caller's information. Meanwhile, the user interface was designed for clarity and intuitive operation. Moreover, Equipped with the TI Aries chipset, HD handset, HD speaker and HD codec (G.722), SIP-T32G IP Phone offers an unrivaled, lifelike audio experience. Furthermore, The T32G Phone supports the security standards which prevent the recording of data, For instance: TLS, SRTP, HTTPS, 802.1x, Open VPN and AES encryption.

Yealink
easy VoIP

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 (1000 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
Basic Hot-desking
XML Browser, XML Idle Screen
Action URL & Active URI
Wallpaper, Screensaver
Color Picture Caller-ID
Theme, Screen Sleep

IP PBX System Integration

Busy lamp field (BLF), BLF list, (BLA)
DND & Forward synchronization
Intercom, Paging
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 optional, TR069 optional
Configuration: browser/phone/auto-provision
Factory configuration customized
Trace package and system log export

Security

Open VPN, 802.1x, VLAN QoS (802.1pq), LLDP
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

2xRJ45 10/100/1000Mbps Ethernet ports
TI Aries chipset
3" TFT-LCD, 400 x 240 pixel, 262K colors
32 keys including 3 programmable keys
5 LEDs: 1 x power, 3 x line, 1 x message
Wall-mountable
1xRJ9 (4P4C) handset port
1xRJ9 (4P4C) headset port
Power adapter: AC 100~240V input and
DC 5V/2A output
Power over Ethernet (IEEE 802.3af)
Power consumption: 2.8-4.35W
Net weight: 0.82KG
Dimension: 209MMx188MMx150MMx41MM
Operating humidity: 10-95%
Operating temperature: -10~50°C

Package Features

Qty/CTN: 10 PCS
N.W/CTN: 11.8KG
G.W/CTN: 13KG
Measurement: 0.083CMB
Carton Meas: 560MMx460MMx250MM

Certifications

RoHS Compliant   CE   FC   ISO 9001

REACH   MIC

BROADSOFT   Asterisk   AVAYA DEVCONNECT

DC5V   Internet   PC

POWER

Internet