Future-proof IP Phone that combines GigE

- Dual-port Gigabit Ethernet, Power over Ethernet
- 4,3" TFT-LCD, 16.7M colors, Intuitive user interface
- TI Aries chipset and TI voice engine
- HD Voice: HD Codec, HD speaker, HD handset
- 6 VoIP accounts, BLF/BLA, IPv6, Open VPN
- Headset, Support Wireless Headset, LCD Expansion module





SIP-T38G 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, The T38G IP Phone offers a brilliant depiction of caller's information, And 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), The T38G IP Phone offers an unrivaled, lifelike audio experience.

Additionally, T38G IP Phone includes full standard feature set of Yealink T28P Version 60

Gigabit Color IP Phone

SIP-T38G

Phone Features

6 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

Supports up to 6 expansion modules(EXP39)
Supports Wireless Headset Adapter(EHS36)

Advanced Features

XML phonebook search/import LDAP phonebook XML Idle Screen Action URL & Active URI Wallpaper, Screensaver Color Picture Caller-ID Theme, Screen Sleep mode

IP PBX System Integration Busy lamp field (BLF), BLF list, (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/PPP0E Bridge/router mode for PC port TFTP/DHCP/PPP0E client Telnet/HTTP/HTTPS server DNS client, NAT/DHCP server

Management

Logout

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

4.3" TFT-LCD, 480 x 272 pixel, 16.7M colors 48 keys including 16 programmable keys

1xRJ9 (4P4C) handset port 1xRJ9 (4P4C) headset port 1XRJ12 (6P6C) EXT port

Power adapter: AC 100~240V input and

DC 5V/2A output

Power over Ethernet (IEEE 802.3af) Power consumption approx.: 4.6W

Net weight: 1.05KG

Dimension: 273x204x42MM Operating humidity: 10~95% Storage temperature: up to 60°C

Package Features

Qty/CTN: 5 PCS N.W/CTN: 8.065KG G.W/CTN: 8.915KG Measurement: 0.062CMB Carton Meas: 580*295*250MM

Certifications











BROADSOFT







