

## *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, 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 a host of telephony features to increase efficiency.

## 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 and EXP38)  
Supports Wireless Headset Adapter(EHS36)

## 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 mode

## IP PBX System Integration

Busy lamp field (BLF), BLF list, (BLA)  
DND & Forward synchronization  
Intercom  
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.711 $\mu$ /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)  
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  
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.: 2.8-4.45W  
Net weight: 0.99KG  
Dimension: 273MM\*200MM\*158MM\*58.5MM  
Operating humidity: 10~95%  
Operating temperature: -10~50°C

## Package Features

Qty/CTN: 5 PCS  
N.W/CTN: 7.5KG  
G.W/CTN: 8.3KG  
Measurement: 0.062CMB  
Carton Meas: 580MMx295MMx250 MM

## Certifications

