

### *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.

## 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.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)  
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

