

Professional IP Phone with 3 lines & HD Voice

Yealink SIP-T23G features intuitive user interface and enhanced functionality which make it easy for people to interact and maximize productivity. Yealink HD technology enables rich, clear, life-like voice communications, outsourced management options, flexible deployment and third-party communications applications. As a cost-effective IP solution, it helps to streamline business processes, delivering a powerful, secure and consistent communication experience for small and large office environments.



SIP-T23G



Key Features and Benefits

HD Audio

Yealink HD Voice refers to the combination of software and hardware design as well as the implementation of wideband technology to maximize the acoustic performance. Coupled with advanced acoustic clarity technology such as full duplex, echo cancellation, adaptive jitter buffer, etc, the SIP-T23G provides clearer more life-like voice communications.

Enhanced Call Management

The SIP-T23G supports vast productivity-enhancing features such as XML Browser, call park, call pickup, BLF, call forward, call transfer and 3-way conferencing, which makes it the natural and obvious efficiency tool for today's busy small and large office environments.

Efficient Installation and Provisioning

Integrated IEEE802.3af Power-over-Ethernet allows easy deployment with centralized powering and backup. The SIP-T23G supports the FTP, TFTP, HTTP and HTTPS protocols for file provisioning and are configured by default to use Trivial File Transfer Protocol (TFTP) and supports AES encrypted XML configuration file.

Highly secure transport and interoperability

The communicator uses SIP over Transport Layer Security (TLS/SSL) to provide service providers with the latest technology for enhanced network security. The range is certified compatible with 3CX, Asterisk and Broadsoft Broadworks, ensuring excellent compatibility with leading soft switch suppliers.

- Yealink HD Voice
- 132x64-pixel graphical LCD
- Dual-port Gigabit Ethernet
- PoE support
- Up to 3 SIP accounts
- Supports Dual-color LEDs
- Headset support
- Wall mountable
- IPv6 support
- Paperless design
- Simple, flexible and secure provisioning options

Audio Features

- > HD voice: HD handset, HD speaker
- > Wideband codec: G.722
- > Narrowband codec: G.711(A/μ), G.729AB, G.726, iLBC
- > DTMF: In-band, Out-of-band (RFC 2833) and SIP INFO
- > Full-duplex hands-free speakerphone with AEC
- > VAD, CNG, AEC, PLC, AJB, AGC

Phone Features

- > 3 SIP accounts
- > One-touch speed dial, redial
- > Call forward, call waiting, call transfer, call hold
- > Call return, group listening, SMS
- > Mute, auto answer, DND
- > Local 3-way conferencing
- > Direct IP call without SIP proxy
- > Ring tone selection/import/delete
- > Hotline, emergency call, off hook hot line
- > Set date time manually or automatically
- > Dial Plan, XML Browser, action URL/URI
- > Enhanced DSS key
- > Integrated Screenshots
- > RTCP-XR, VQ-RTCPXR

Directory

- > Local phonebook up to 1000 entries
- > Call history: dialed/received/missed/forwarded
- > Black list
- > XML/LDAP remote phonebook
- > Intelligent search method
- > Phonebook search/import/export

IP-PBX Features

- > Busy Lamp Field (BLF)
- > Bridged Line Appearance (BLA)
- > Anonymous call, anonymous call rejection
- > Voice mail, call park, call pickup
- > Intercom, paging, music on hold
- > Hot-desking, emergency call
- > Message Waiting Indicator (MWI)
- > Call completion, call recording

Display and Indicator

- > 132x64-pixel graphical LCD with backlight
- > LED for call and message waiting indication
- > Dual color illuminated LEDs for line status info
- > Intuitive user interface with icons and soft keys
- > National language selection
- > Caller ID with name and number

Feature keys

- > 3 line keys with LED
- > 5 features keys: message, headset, redial, hands-free speakerphone, transfer
- > 6 navigation keys
- > Volume control keys

Interface

- > Dual-port Gigabit Ethernet
- > Power over Ethernet (IEEE 802.3af), Class 2
- > 1xRJ9 (4P4C) handset port
- > 1xRJ9 (4P4C) headset port

Other Physical Features

- > Wall mountable
- > External universal AC adapter (optional): AC 100~240V input and DC 5V/600mA output
- > Power consumption (PSU): 0.72W-1.56W
- > Power consumption (PoE): 0.96W-1.98W
- > Dimension(W*D*H*T): 209mm*188mm*150mm*41mm
- > Operating humidity: 10~95%
- > Operating temperature: -10~50°C

Management

- > Configuration: browser/phone/auto-provision
- > Auto provision via FTP/TFTP/HTTP/HTTPS for mass deploy
- > Auto-provision with PnP
- > Zero-sp-touch, TR-069
- > Phone lock for personal privacy protection
- > Reset to factory, reboot
- > Package tracing export, system log

Network and Security

- > SIP v1 (RFC2543), v2 (RFC3261)
- > Call server redundancy supported
- > NAT transverse: STUN mode
- > Proxy mode and peer-to-peer SIP link mode
- > IP assignment: static/DHCP
- > HTTP/HTTPS web server
- > Time and date synchronization using SNTP
- > UDP/TCP/DNS-SRV(RFC 3263)
- > QoS: 802.1p/Q tagging (VLAN), Layer 3 ToS DSCP
- > SRTP for voice
- > Transport Layer Security (TLS)
- > HTTPS certificate manager
- > AES encryption for configuration file
- > Digest authentication using MD5/MD5-session
- > OpenVPN, IEEE802.1X
- > IPv6

Package Features

- > Qty/CTN: 10 PCS
- > N.W/CTN: 11.8 kg
- > G.W/CTN: 13.0 kg
- > Giftbox size: 215mm*200mm*118mm
- > Carton Meas: 615mm*436mm*208mm

Certifications

