

# Yealink SIP-T23G Professional IP Phone with PoE, PSU

57,900 AMD

## Audio Features

- > HD voice: HD handset, HD speaker
- > Wideband codec: G.722
- > Narrowband codec: G.711(A/Î¼), G.729AB,G.726, iLBC
- > DTMF: In-band, Out-of-band(RFC 2833) and SIPÂ INFOÂ
- > Full-duplex hands-free speakerphone with AEC
  
- > VAD, CNG, AEC, PLC, AJB, AGC

## Phone Features

- > 3 SIP accounts
- > Call hold, mute, DND
- > One-touch speed dial, hotlineÂ
- > Call forward, call waiting, call transfer
- > Group listening, SMS
- > Redial, call return, auto answer
- > Local 3-way conferencing
- > Direct IP call without SIP proxy
- > Ring tone selection/import/delete
- > Set date time manually or automatically
- > Dial plan, off hook hot line
- > XML Browser, Action URL/URI
- > Integrated Screenshots
- > RTCP-XR

## Directory

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

## IP-PBX Features

- > Busy Lamp Field (BLF)
- > Bridged Line Apperance(BLA)

- > Anonymous call, anonymous call rejection
- > Hot-desking, emergency call
- > Message Waiting Indicator (MWI)
- > Voice mail, call park, call pickup
- > Intercom, paging, music on hold
- > Call completion, call recording

#### Display and Indicator

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

#### Feature keys

- > 3 line keys with LED
- > 5 feature keys: message, headset, redial, transfer, hands-free speakerphone
- > 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): 0.72-1.56W
- > Power consumption (PoE): 0.96-1.98W
- > 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)
- > 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-sess
- > OpenVPN, IEEE802.1X
- > IPv6

#### Package Features

- > Qty/CTN: 10 PCS
- > G.W/CTN: 12.6kg
- > Giftbox size: 215mm\*200mm\*118mm
- > Carton Meas: 615mm\*436mm\*208mm

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Audio

Features

>

HD

voice:

HD

handset,

HD

speaker

>

Wideband

codec:

G.722

>

Narrowband

codec:

G.711(AÍ¼),

G.729AB,G.726,

iLBC

>

DTMF:

In-band,

Out-of-band(RFC

2833)

and

SIPÂ

INFOÂ

>

Full-duplex

hands-free

speakerphone

with

AEC

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VAD,

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