

Yealink[®] VVX[®] T40G Gigabit IP Phone

Yealink

Gigabit IP Phone with three lines and HD voice

Yealink

(KyleWu

T40G

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Yealink SIP-T40G is a feature-rich SIP phone that enhances daily interoperability and unifed communications for busy managers. Designed with a revolutionary new appearance, it supports flexible and secure provisioning, and is intuitively designed with ease-of-use in mind.

Yealink Optima HD voice

HD

- 2.3" 132x64-pixel graphical LCD with backlight
- **Dual-port Gigabit Ethernet** •
- PoE support

Yealink

Audio Features

- HD voice: HD handset, HD speaker
- Wideband codec: G.722
- Narrowband codec: G.711(A/μ), G.729AB
- 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 VoIP accounts
- One-touch speed dial, redial
- Call forward, call waiting
- Call transfer, call hold
- Call return, group listening, SMS
- Mute, auto answer, DND
- 3-way conference call
- Direct IP call without SIP proxy
- Ring tone selection/import/delete
- Hotline, emergency call
- Set date & time manually or automatically
- Dial Plan
- XML Browser
- Action URL/URI
- RTCP-XR (RFC3611), VQ-RTCPXR (RFC6035)
- Enhanced DSS Key

Directory

- Local phonebook up to 1000 entries
- Black listXML/LDAP remote phonebook
- Smart dialing
- Phonebook search/import/export
- Call history: dialed/received/missed/ forwarded

IP-PBX Features

- Busy Lamp Field (BLF)
- Anonymous call, anonymous call rejection
- Voicemail
- Message Waiting Indicator (MWI)
- Call park, call pickup
- Intercom, paging, music on hold
- Call recording
- Flexible seating
- Executive and Assistant
- Centralized call recording
- Visual Voice Mail
- Display and Indicator
- 2.3" 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
- Multilingual user interface
- Caller ID with name and number
- Power saving

Feature keys

- 3 line keys with LED
- 5 features keys: message, headset, mute,
- redial, hands-free speakerphone
- 4 context-sensitive "soft" keys
- 6 navigation keys
- 2 volume control keys
- Mute key
- Headset key
- Hands-free speakerphone key

Interface

- Dual-port Gigabit Ethernet
- 1 x RJ9 (4P4C) handset port
- 1 x RJ12 (6P6C) EHS port
- Power over Ethernet (IEEE 802.3af), Class 2

Other Physical Features

- Color: Classic Grey
- Stand with 2 adjustable angles
- Wall mountable
- External Yealink AC adapter (optional):
- AC 100~240V input and DC
- 5V/600mA output
- Power consumption (PSU): 0.9-1.3W
- Power consumption (PoE): 1.1-1.6W
- Dimension (W*D*H*T):
- 212mm*189mm*175mm*54mm
- Operating humidity: 10~95%
- Operating temperature: -10~50°C (+14~122°C)

Management

- Configuration: browser/phone/autoprovision
- Auto provision via FTP/TFTP/HTTP/HTTPS for mass deploy
- Auto-provision with PnP
- BroadSoft device management
- 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 traversal: STUN mode

T40G Gigabit IP Phone Product Specifications

- 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

Transport Layer Security (TLS)

AES encryption for configuration file

HTTPS certificate manager

Digest authentication

OpenVPN, IEEE802.1X

LLDP/CDP/DHCP VLAN

- Yealink SIP-T40G IP phone

- Handset with handset cord

- Power Adapter (Optional)

Giftbox size: 246mm*223mm*120mm

Carton Meas: 620mm*233mm*258mm

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Asterisk[®]

DSCP

IPv6

ICF

Package Features

Package content:

- Ethernet Cable

- Quick Start Guide

Qty/CTN: 5 PCS

N.W/CTN: 5.7 kg

G.W/CTN: 6.4 kg

metaswitch

- Stand

Compliance

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SRTP for voice