

SIP-W52P

Up to 5 VoIP accounts and 4 simultaneous calls Cat-iq2.0, HD Voice (G.722), PoE Class 1 Expandable up to 5 DECT cordless handsets 1.8"128x160 TFT colour display 10 hours talk time, 100 hours standby time Range up to 50m indoors and 300m outdoors



Flexible cordless VoIP solution for small business

Yealink IP DECT SIP-W52P is a SIP Cordless Phone that's designed for small business users who are looking for immediate cost savings with scalability and efficiency.

Handset 1

Options

09:56AM

× 03

The high-quality TFT colour display provides a clear view from all angles and the high definition speakerphone ensures clear handsfree communication. The SIP-W52P supports 4 simultaneous calls per base and offers professional features including intercom, transfer, 3-way conferencing and speed dial.

It also works with well known brand platforms including Broadsoft, Asterisk and 3CX and supports quick and easy configuration.



SIP-W52P

Phone Features

5 VoIP accounts, up to 5 handsets Supports 4 simultaneous calls per base

Call hold, Call transfer (Blind/Attend/ Semi-Attend)

Divert, Switching between calls 3-way conference, Call waiting Caller ID display, Redial, Mute, DND Call forward (Always/Busy/No Answer) Speed dial, Voicemail, Silence Message Waiting Indication (MWI) LED Phonebook:500 entries, Black list Call history(Outgoing/Missed/Accepted) Phonebook search / import / export Keypad lock, Emergency call 6 Ringtones Energy-saving ECO mode

Advanced Features

Multiple languages

Remote phonebook Direct IP call without SIP proxy

Screen Saver, Two colour schemes

IP PBX System Integration

Music on hold, Dial now, Call return Paging, Intercom, Auto answer Anonymous call, Anonymous call rejection

Dial Plan ,Emergency call

Management

Unified Auto provisioning template Auto-provision via FTP / TFTP / HTTP / **HTTPS**

Auto-provision with PnP Configuration: browser / phone / autoprovision

Factory configuration customized Trace package and system log export

Voice and Codecs Features

Full-duplex speakerphone with AEC Receiver volume control Ring tone selection Ringer volume Key sounds – click, DTMF and special

tones Wideband codec: G.722 Narrowband codec: G.711µ/A, G.723, G.726, G.729AB, iLBC VAD, CNG, AEC, PLC, AJB

Network Features

SIP v1 (RFC2543), v2 (RFC3261) SNTP/NTP VLAN(802.1Q and 802.1P) STUN Client (NAT Traversal) DTMF: In-Band, RFC2833, SIP Info UDP,TCP,TLS IP Assignment: Static/DHCP/PPPoE 802.1x, LLDP, PPPoE Open VPN

Security

VLAN QoS (802.1pq) Transport Layer Security (TLS) HTTPS (server / client) SRTP (RFC3711) Digest authentication using MD5 Secure configuration file via AES encryption Admin / Var / User 3-level configuration

Password protection

Connectors

mode

DECT/Cat-iq2.0 1 x RJ45 10/100M Ethernet port Power over Ethernet (IEEE 802.3af) Headphone jack (2.5 mm)

Physical Features

1 x Handset, 1 x Base, 1 x Belt Clip, 1 x Cable, 2 x Guide

Range: 50m indoor, 300m outdoor Coverage be extended using repeaters (Pendina)

10 hours talk time, 100 hours standby time

1.8"TFT colour display: 128x160 pixels, LCD backlit, Key backlit

12 key numerical keypad, 5 navigation keys

2 softkeys, 2 hook keys, 1 Mute key 1 speaker key,1 Message key

3 LEDs on Base: 1 x power, 1 x Network, 1 x Call

Base station: DC 5V / 600mA Output Charger: DC 5V / 600mA Output 2 x Ni-MH AAA :1.2V/800mA

Phone size: 144mm x 50mm x 24mm Base station size: 153mm x 108mm x 45mm

Operating humidity: 10~95%

Certifications

WEEE Compliant



ROHS Compliant





REACH









