

CP860

Full duplex HD IP Conference Phone

Power over Ethernet

10 foot range and 360 degree voice pickup

5 way conference

USB call recording

PC audio connection

Optional expansion microphones



Optima
HD Voice



5-way
conferencing



Call
recording



Mobile phone
/ PC port



Expansion
microphones kit

HD SIP Conference Phone enhances collaboration and improves productivity

The Yealink CP860 IP Conference phone is the perfect choice for any business needing to communicate with dispersed teams, customers, partners and suppliers. Ideal for small and medium-sized conference rooms, the phone can be used for up to 16 people when using the optional expansion microphones. The Yealink CP860 benefits from optima HD technology, 3 built-in microphones, full duplex and acoustic echo cancelling providing a rich and clear audio experience. In addition the CP860 supports call recording when utilizing a USB flash drive.

CP860

CP860 Accessories

Microphone Expansion Kit

Optima HD Voice
Full duplex technology
Echo cancellation

10 foot range and 360 degree voice pickup
Mute function via touchpad



Audio Features

Optima HD voice
10 foot microphone pickup
Built in 3 microphone array, 360 degree voice pickup
112 mm diameter and 2.7w speaker
Full-duplex speakerphone with AEC
Echo cancellation tail length is up to 400ms
Reverberation elimination
Background noise suppression
Codecs: G.722, G.711(A/μ), G.729AB, iLBC
DTMF: In-band, Out-of-band(RFC 2833) and SIP INFO
VAD, CNG, PLC, AJB, AGC

Phone Features

1 VoIP account
Call hold, mute, DND
Call recording, hotline
5-way conference calls
Redial, call waiting, emergency call
Call forward, call transfer, call return, dial plan
Ring tone selection/import/delete
Set date time manually or automatically
Volume adjustment

Directory

Local phonebook up to 1000 entries
XML/LDAP remote phonebook
Intelligent search method

Phonebook search/import/export
Call history: dialled/received/missed/forwarded
Black list

IP-PBX Features

Intercom, multicast paging
Anonymous call, anonymous call rejection
Voice mail, distinctive ringtone, call pickup

Display

192x64-pixel graphical LCD with backlight
24 total keys: 4 softkey, off-hook key, on-hook key, redial, mute, volume key, 3 navigation keys
Phone lock
National language selection
Caller ID with name, number
LCD display logo

Interface

1xRJ45 10/100M Ethernet port
Power over Ethernet (IEEE 802.3af), class 0
2xEX mic port
1xUSB2.0 port
1xmobile phone/PC port: 3.5mm
1xSecurity lock port

Network and security

SIP v1 (RFC2543), v2 (RFC3261)
SIP server redundancy supported
IPv4/IPv6
NAT traverse: 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 encryption
Transport Layer Security (TLS)
HTTPS certificate manager
AES encryption for configuration file
Digest authentication using MD5/MD5-session
OpenVPN, IEEE802.1X

Management

Setup wizard
Configuration: browser/phone/auto-provision
Auto provision via FTP/TFTP/HTTP/HTTPS for mass deploy
Auto-provision with PnP
Zero-sp-touch, TR-069, SNMP
Reset to factory, reboot
Package tracing export, system log

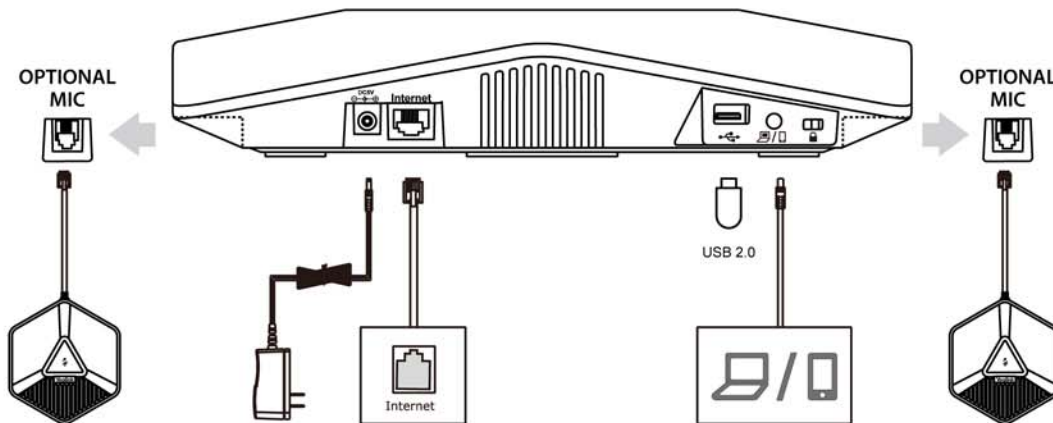
Other physical features

4 context-sensitive "soft" keys
External universal AC adapter (optional): AC 100~240V input and DC 5V/2A Output
Power consumption (PSU): 2.8-5.6W
Power consumption (PoE): 4.0-8.2W
Dimension(W*D*H): 320mm*260m*52mm
Operating humidity: 10~95%
Storage temperature: -10~50°C

Certifications



...and many more



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