

Dual Mode IP Phone: SIP + Teams /Zoom

UNIVOIS

Connect Your Phones



OFFICE DUAL MODE IP PHONE

Univois UC400

Gigabit Colour Screen IP Phone

18 SIP Lines

Key Features:

- 4.5" 480x854 pixels, colour display
- 18 SIP accounts
- 90 DSS keys
- HD voice
- Full-duplex hands-free speakerphone
- 2 x 10/100/1000M Ethernet ports
- Power over Ethernet (PoE)
- Electronic Hook Switch (EHS)
- Built-in USB interface
- Support Teams/Zoom
- Support PC USB Audio
- Supports up to two Univois 400IP-AC expansion modules, for up to 72 additional DSS Keys
- LCD supports multi-languages: English, Chinese, French, Italian, Portuguese, Portuguese-Brazil, Russian, Spanish, Turkish, Hebrew, Chinese (Traditional), Korean, Netherlands and Slovak
- Web supports multi-languages: English, Chinese, French, Italian, Portuguese, Russian, Spanish, Turkish and Chinese (Traditional)

PHONE FEATURES

- 18 SIP accounts, Hotline
- Support Teams/Zoom
- Call hold, Call waiting, Call forward, Call return
- Call transfer (blind/busy/ask), Incoming call transfer
- Caller ID display, Redial, Mute, DND
- Auto Answer, three-way conferencing
- Speed dial, SMS, Voicemail Message Waiting Indication LED
- Tone scheme, Volume control
- Direct IP call without SIP proxy
- Ringtone selection/import/delete
- Blacklist
- Handsfree indicator
- Call history: dialled/received/missed
- Multi-language
- Programmable Soft key

ADVANCED FEATURES

- XML phonebook input & output
- Up to 2000 entry enterprise phonebook
- Up to 1000 entry personal phonebook
- LDAP phonebook
- Functions customizable keys
- Supports up to 90 paperless DSS keys
- Electronic Hook Switch (EHS)
- Supports USB Charging

IP PBX SYSTEM INTEGRATION

- Busy lamp field (BLF), BLF list
- Bridged line appearance (BLA/SLA)
- DND & Forward synchronization
- Intercom, Paging, Music on hold
- Call pickup, Dial Plan, Call recording, Anonymous call, Anonymous call rejection
- Network conference, Distinctive ringtone

SECURITY

- LLDP, VLAN, VPN (L2TP), Open VPN, 802.1X
- Transport Layer Security (TLS)
- Digest authentication using MD5/MD5-sess
- Secure configuration file via AES encryption
- Phone lock for personal privacy protection
- Admin/User 2-level configuration mode

CODEC AND VOICE FEATURES

- Wideband Codec: G.722, Opus
- Narrowband codec: G.711 μ /A, G.726, G.729AB, iLBC
- VAD, CNG, AEC, AGC
- Full-duplex hands-free speakerphone

NETWORK FEATURES

- SIP v1 (RFC2543), v2 (RFC3261)
- DNS SRV (RFC3263), QoS (802.1pq)
- NAT Traversal: STUN mode
- DTMF: In-Band, RFC2833, SIP Info, Auto
- HTTP Web Management
- IP Assignment: Static/DHCP/PPPoE Bridge/Router mode for PC port
- TFTP/DHCP/PPPoE client
- DNS client, NAT/DHCP server

MANAGEMENT

- Auto-provision via FTP/TFTP/HTTP/HTTPS
- Configuration: browser/phone/auto-provision
- Trace package and system log export, TR069

OPTIONAL FEATURES

- HTTPS Web management, IPV6

PHYSICAL FEATURES

- 4.5" 480x854 pixels, colour display
- 49 keys, 21 LEDs
- Desk stand
- 1 x RJ-9 handset port
- 1 x RJ-9 headset port
- 1 x USB port
- 2 x RJ45 10/100/1000M Ethernet ports
- DC 12V/1A input
- Power over Ethernet, IEEE 802.3af, class 0
- Power consumption: 2.5-3.5W
- Dimension: 245x178x64mm
- Operating humidity: 10-90%
- Storage temperature: up to 60°C



To obtain service or information

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Product specification is subject to change without prior notice

EU Version: 4