



UT-H2U Compact SIP IP Phone

Tiny and stylish, Uniteli UT-H2U compact IP phone possesses brand features including 2 SIP lines, 10 speed dial keys, 1 programmable DSS key and a HD speaker on handset. It's support the install mode of desktop or wall-mounted, and power supply mode of PoE or power adapter.

The Uniteli UT-H2U is more than a hotel phone and also can be applied in multiple scenarios, such as school, hospital, supermarket and residence.

Highlights

- ✓ Delicate and fashion outlook.
- ✓ Easy installation and configuration.
- ✓ Cost-effective model & Environmental friendly design.
- ✓ 10 speed dial keys, long press numeric keypad to achieve speed dial
- ✓ 1 programmable DSS key.
- ✓ 10/100Mbps Ethernet, support PoE power supply.
- ✓ Support the install mode of desktop or wall-mounted
- ✓ Compatible with major platforms: Asterisk, Broadsoft, 3CX, Metaswitch, Elastix, Avaya etc.



Generic

- 2 SIP Lines
- HD voice
- PoE enabled
- Intelligent DSS keys
- Handset / hands-free mode
- Desktop / wall-mounted
- Optional external power supply

Phone Functions

- Call logs (In/Out/Missed, Max support 600 records in the web)
- Blocked list / allowed list
- Voice message waiting indication (VMWI)
- Programmable DSS/soft keys
- Network time synchronization
- Action URL / Active URI
- uaCSTA

Call Functions

- Call out / answer / reject
- Mute / unmute (microphone)
- Call hold / resume
- Call waiting
- Intercom
- Speed dial
- Anonymous call (hide caller ID)
- Redial
- Auto-answering
- Voice message (with server)

- 3-way conference (Passive invite access)
- Hot line
- Hot desking

Audio

- HD voice microphone/speaker (handset/hands-free, 0 ~ 7KHz frequency response)
- HAC handset
- Wideband ADC/DAC 16KHz sampling
- Narrowband codec: G.711a/u, G.723.1, G.726, G.729A/B, iLBC
- Wideband codec: G.722, Opus
- Full-duplex Acoustic Echo Cancellation (AEC)
- Voice Activity Detection (VAD) / Comfort Noise Generation (CNG) / Background Noise Estimation (BNE) / Noise Reduction (NR) / Automatic Gain Control (AGC)
- Packet Loss Concealment (PLC)
- Dynamic adaptive jitter buffer
- DTMF: In-band, Out-of-Band (RFC2833/ SIP INFO)
- VQM voice quality monitoring

Network

- Physical: 10/100Mbps ethernet
- IP Mode: IPv4/IPv6/IPv4&IPv6
- IP Configuration: Static IP / DHCP / PPPoE
- Network Access Control: 802.1x
- VPN: L2TP / OpenVPN
- VLAN
- LLDP
- CDP

- QoS
- RTCP-XR (RFC3611), VQ-RTCPXR (RFC6035)

Protocols

- SIP2.0 over UDP/TCP/TLS
- RTP/RTCP/SRTP
- STUN
- DHCP
- IPv6
- LLDP
- CDP
- PPPoE
- 802.1x
- L2TP
- OpenVPN
- SNTP/NTP
- FTP/TFTP
- HTTP/HTTPS
- TR-069
- AES128 & AES256

Deployment & Management

- Auto-provisioning via FTP/TFTP/HTTP/HTTPS/DHCP OPT66/SIP PNP/TR-069
- Web management portal
- Web-based packet dump
- Configuration export / import
- Firmware upgrade
- Syslog

Physical Specifications

- Keypad: 19 keys, including
 - 12 Standard Phone Digits keys(0-9: Speed dial configuration on webpage)
 - 3 Function keys (Hands-free , Redial ,Hold)
 - 3 Volume Control keys, Up/Down/Mute(with red color LED)
 - 1 DSS key
- HD handset (RJ11) x1
- RJ11 handset cord x1
- Status indicator light x1 (two-color LED)
- RJ11 port x1: handset x1
- RJ45 port x1: network x1
- DC port x1: Φ 5.5x2.1x10mm
- DC power input: DC5V / 2A or PoE
- PoE level: Class 1
- Working temperature: 0~45℃
- Working humidity: 10~95%
- Installation: Desktop / Wall-mounted
- Color: black/white
- Device dimensions: 55.8x215x57.8 mm
- Gift box dimensions: 224x75x62 mm
- Outer CTN dimensions: 335x323x243 mm (20 PCS)

The above specifications may be updated in the future without prior notice. All hardware/software/physical features should be based on the final shipped products.

Compatibility List:

								
								