

Two Lines Business IP Phone

UTT-290 Datasheet



UTT-290 is a highly innovative based SIP VoIP phone, it can perfectly satisfy with all kinds of the businessman communication. UTT-290 integrates with 132x64 graphic LCD, a large resolution display screen, elegant and intuitive user interface, which indicate you can enjoy perfect user experience. UTT-290 offers high audio quality HD handset, HD speaker and HD codec (G.722), with a rich set of business communication features that can definitely improve working efficiency and sense. UTT-290 is highly suitable for SMEs, home Office and ISP applications.

Login

Username: root

Password: root

Key Features

- 132*64 Graphic LCD.
- HD Voice: HD Codec, HD Speaker, HD Handset.
- 2 level adjustable bracket and wall-mountable.

- Two SIP accounts and support three-way conference, SMS.
- 2xLAN, PoE (option), 3.5mm and RJ9Headset.
- USB port only support DC 5V output feature.
- XML/LDA, Contact Search.
- Auto-provision, HTTP/TFTP/FTP, Tr069.
- Built-in languages: Simple Chinese, English, French, Italian, Polish, Portuguese, Russian, Spanish, Turkish etc.

Phone Features

- 2 SIP accounts, Hotline
- Call hold, Call waiting, Call forward, Call return
- Call transfer (blind/busy/ask)
- Caller ID display, Redial, Mute, DND
- Auto-answer, 3-way conferencing
- Speed dial, SMS, Voicemail
- Message Waiting Indication LED
- Tone scheme, Volume control
- Direct IP call without SIP proxy
- Ring tone selection/import/delete
- Phonebook (Personal Phone Book Search/Input/Output (300 entries),
- Black list, Hand-free indicator.
- Call history: dialed/received/missed (100 -entries); One-key back (C)
- Multi-Language (2aB/abc/ABC/123)
- Soft keys programmable, Supports PC control
- Support switching SIP server & Second server

Advanced Features

- XML Phone Book search and input & output
- Enterprise phone book(800 entries)
- LDAP phone book
- Functions customizable keys

IP PBX System Integration

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

Security

- 802.1x, VLAN QoS (802.1pq)

- Transport Layer Security (TLS)
- Digest authentication using MD5/MD5-sess
- Secure configuration file via AES encryption
- Phone lock for personal privacy protection
- Admin/VAR/User 3-level configuration mode

Network Features

- SIP v1 (RFC2543), v2 (RFC3261)
- SIP connection: Proxy mode & Peering Points mode
- DNS SRV (RFC3263)
- Redundant server support
- NAT Traversal: STUN mode
- DTMF: In-Band, RFC2833, SIP Info, Auto
- HTTP/HTTPS Web Management
- IP Assignment: Static/DHCP/PPPoE
- Bridge/Router mode for PC port
- TFTP/DHCP/PPPoE client
- Telnet/HTTP/HTTPS server
- DNS client, NAT/DHCP server

Management

- Auto-provision via FTP/TFTP/HTTP/HTTPS
- SNMP V1/2 , TR069
- Configuration: browser/phone/auto-provision
- Trace package and system log export

Codes and Voice Features

- Wideband Codec: G.722
- Narrowband codec: G.711 μ A, G.723.1
- G.726, G.729AB, VAD, CNG, AEC, AGC
- Full-duplex,

Physical Features

- 132x64 graphic LCD
- 33 keys including 2 programmable keys
- 1x3.5mm handset port, 1xRJ9 headset port
- 2xRJ45 10/100M Ethernet ports
- 2 level adjustable bracket and wall-mountable
- Power adapter: AC 100~240V input and
- DC 5V/1A output
- USB port only support DC 5V output feature.
- Power over Ethernet , IEEE 802.3af, class 0

- Power consumption: 1.4-2.6W
- Net weight: 0.80KG
- Dimension: 185x200x90MM
- Operating humidity: 10~95%
- Storage temperature: up to 60 °C

Package Features

- Qty/CTN: 10 PCS
- N.W/CTN:12.50 KGS
- G.W/CTN: 13.50KGS
- Measurement: 0.068 CBM (0.5% more or less)
- Carton Meas: 500x480x320MM

Certifications

- RoHS
- CE
- FCC
- ISO9001

Compatible with the platform

- AVAYA
- Alcatel-Lucent
- Asterisk
- Broadsoft
- Elastix
- Genesys
- Huawei
- MetaSWITCH
- NEC
- Yearstar
- ZTE
- 3CX