

ESCENE CC800V2 is one of IP Phone in the Call Center. It has has the unique style, good utility, clear voice etc feature. Cooperating with communication platform to finish strong phone functions, such as: call transfer, hotline function (immediately/delay), a key cancellation and registration, a key automatic response, etc.

Key Features

- One key enable or disable accounts register function.
- One key enable or disable auto-answer function.
- One key change the ringing type.
- Multi-language, e.g. Chinese, English, Russian, French etc.
- Two SIP accounts and support three-way conference, SMS.
- 2xLAN, PoE, RJ9Headset.
- 4 programmable keys.
- XML/LDA, BLF/BLA
- Auto-provision, HTTP/TFTP/FTP.
- Light of status.

CC800V2 Technical Datasheet

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 Black list, Hand-free indicator. Call history: dialed/received/missed (50 -entries); Multi-Language Soft keys programmable, Supports PC control

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

Compatible with the platform

Codes and Voice Features

Wideband Codec: G.722 Narrowband codec: G.711µ/A, G.726, G.729AB, iLBC VAD, CNG, AEC, AGC.Full-duplex, Security LLDP, VLAN QoS (802.1pq), VPN(L2TP)* Transport Layer Security (TLS)* Digest authentication using MD5/MD5-sess Secure configuration file via AES encryption * Encryption for configuration file* Phone lock for personal privacy protection Admin/User 2-level configuration mode, LLDP, CDP **Network Features** SIP v1 (RFC2543), v2 (RFC3261) SIP connection: Proxy mode & Peering Points mode; SRTP voice encryption(RFC3711) Redundant server support NAT Traversal: STUN mode DTMF: In-Band, RFC2833, SIP Info, Auto HTTP Web Management IP Assignment: Static/DHCP/PPPoE TFTP/DHCP/PPPoE client Telnet/HTTP/HTTPS server DNS client, NAT/DHCP server Management Auto provision via FTP/TFTP/HTTP/HTTPS for mass deploy; Configuration: browser/phone/auto-provision

Trace package and system log export

Optional Feature

TR069, IPv6, HTTPS web management, 802.1x, OPEN VPN

Physical Features

128x64 graphic LCD 25 keys including 2 programmable keys 6 LEDs: 1* headset light, 2* accounts lights, 3*function lights (Answer Key\Ringing\Mute) 1* RJ-9 handset port 1*RJ-9 headset port 1*3.5mm PC headset port 2xRJ45 10/100M Ethernet ports Power adapter: AC 100~240V input and DC 5V/1A output Power over Ethernet .IEEE 802.3af.class 0 Power consumption: 2.5-3.5W Net weight: 0.342KG Dimension: 162x105x62MM Operating humidity: 10~95% Storage temperature: up to 60°C

Package Features

Qty/CTN: 20 PCS N.W/CTN: 6.84KG G.W/CTN: 10.08KG Carton Meas: 285*460*300MM

Certifications





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*these features are not available and blocked in firmware for products which are supplied to Russian Federation and EAC countries