



ZC200

Call Center IP Phone

Lanizer ZC200 is a professional call center IP phones with intuitive, clean design and rich features at a cost-effective price.

Highlights

- ✓ Professional IP phone customized for call center
- ✓ 2 SIP Lines, 3-way conference
- ✓ 2.8" (320x240) color-screen
- ✓ Support call center headset (RJ9 port, 3.5mm port)
- ✓ Electronic Hook Switch (EHS) support for wireless headset
- ✓ LED function Keys for Mute/Hold/Call Forwarding/Headset
- ✓ Support off-hook/on-hook by pedal switch
- ✓ Support LDAP, XML Phonebook
- ✓ 10/100Mbps Ethernet, PoE integrated
- ✓ Compatible with major platforms: Asterisk, Broadsoft, 3CX, Metaswitch, Elastix, Avaya etc.



Model	Screen	Network	PoE	Power adapter
ZC200	2.8" color-screen	10/100Mbps	✓	—

Generic

- 2 SIP Lines
- HD Voice
- PoE Enabled
- Headset mode
- Optional External Power Supply

Phone Features

- Local Phonebook (500 entries)
- Remote Phonebook (XML/LDAP, 500 entries)
- Call logs (In/out/missed, 600 entries)
- Black/White List Call Filtering
- Screen saver
- Voice Message Waiting Indication (VMWI)
- Programmable Soft keys
- Network Time Synchronization
- Support Plantronics wireless headset (Through Plantronics APD-80 EHS Cable)
- Support Jabra wireless headset (Through EHS Cable)
- Support Recording (Through Server Recording)
- Action URL / Active URI
- uaCSTA

all Features

- Call out / Answer / Reject
- Mute / Unmute (Microphone)
- Call Hold / Resume
- Call Waiting
- Intercom
- Caller ID Display
- Anonymous Call (Hide Caller ID)
- Call Forwarding (Always/Busy/No Answer)
- Call Transfer (Attended/Unattended)
- Call Parking/Pick-up (Depending on server)
- Redial
- Do-Not-Disturb
- Auto-Answering
- Voice Message (On server)
- 3-way Conference
- Hot Line
- Hot desking

Audio

- Wideband ADC/DAC 16KHz Sampling
- Narrowband Codec: G.711a/u, G.723.1, G.726-32K, G.729AB
- Wideband Codec: G.722
- Full-duplex Acoustic Echo Canceller (AEC)
- Voice Activity Detection (VAD) / Comfort Noise Generation (CNG) / Background Noise Estimation (BNE) / Noise Reduction (NR)
- Packet Loss Concealment (PLC)

- Dynamic Adaptive Jitter Buffer up to 300ms
- DTMF: In-band, Out-of-Band – DTMF-Relay(RFC2833) / SIP INFO

Networking

- Physical: 10/100Mbps Ethernet, dual bridged port for PC bypass
- IP Configuration: Static / DHCP / PPPoE
- Network Access Control: 802.1x
- VPN: L2TP / OpenVPN
- VLAN
- LLDP
- QoS
- RTCP-XR(RFC3611), VQ-RTCPXR(RFC6035)

Protocols

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

Deployment & Maintenance

- Auto-Provisioning via FTP/TFTP/HTTP/HTTPS/DHCP OPT66/SIP PNP/TR-069
- Web Management Portal
- Web-based Packet Dump
- Configuration Export / Import

- Phonebook Import/Export

- Firmware Upgrade
- Syslog

Physical Specifications

- LCD x1:
 - 2.8 inch (320x240) color-screen
- Keypad: 23 keys, including
 - 4 Soft-keys
 - 4 LED Function keys (Mute, Hold, Auto Answer, Headset)
 - 12 Standard Phone Digits keys
 - 2 Volume Control keys, Lower/Increase(Left/Right)
 - 1 Volume Select key, (Page Down)
- 1.5M CAT5 Ethernet Cable x1
- Earphone bracket x1
- Back Rack x1
- 3.5mm Port x2: Headset x1, Foot pedal x1
- RJ9 Port x2: Headset x1, Monitor earphone x1
- RJ45 Port x2: Network x1, PC x1 (Bridged to Network)
- DC Power Input: 5V/0.6A
- Power Consumption: Idle 0.9~1.6W, Peak 1.6~2.5W
- Working Temperature: 0~45°C
- Working Humidity: 10~95%
- Installation: Desktop Stand
- Color: Black
- Device Dimensions
 - Do not contain Earphone bracket: 100.8x147x107.4mm
 - Contain Earphone bracket: 100.8x155.4x246mm
- Gift Box Dimensions:
 - X2P: 196.98x109.98x65mm
- Outer CTN Dimensions:
 - X2P: 455x340x220mm (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:

