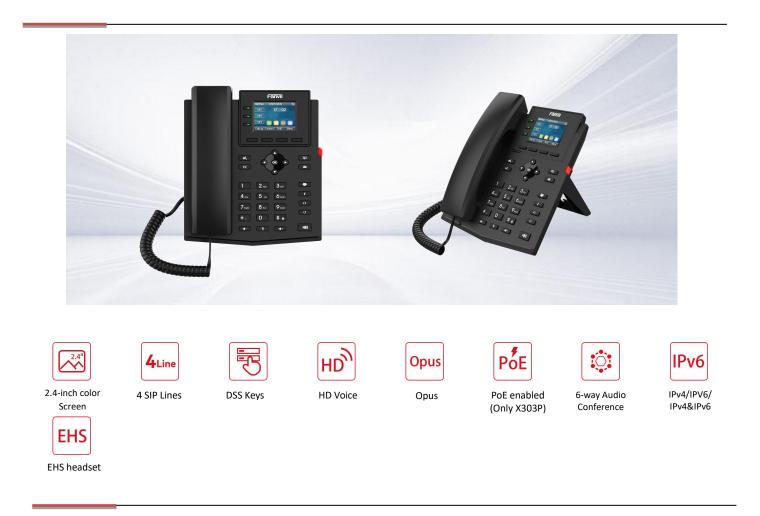


xz

X303/X303P Entry Level IP Phone

X303/X303P is the cost-effective color screen SIP phone with high performance specially for enterprise. It features a 2.4-inch 320x240 pixel color screen, supporting 4 SIP lines and 6-way audio conference to render a user-friendly experience. X303/X303P has rich functions such as HD audio, dual 10/100 Mbps network ports and integrated PoE (only X303P), providing an efficient and high-quality business communication experience for enterprises.



<u>Highlights</u>

- \checkmark 2.4-inch 320x240 color screen.
- Support audio codec G.722 and Opus, HD audio on speakerphone and handset.
- ✓ Support 6-way audio conference.
- \checkmark Support EHS wireless headset.
- □ Dual 10/100 Mbps ports, integrated PoE(only X303P).
- \checkmark Two installation methods of desktop stand and wall mountable.

Generic

- 4 SIP Lines
- HD Voice
- PoE Enabled (only X303P)
- 2.4-inch color screen
- Handset / Hands-free / Headset mode
- Desktop Stand / Wall mountable • (optional)
- **Optional External Power Supply**

Phone Functions

- Local Phonebook (1000 entries)
- Remote Phonebook (XML/LDAP, 1000 • entries)
- Call logs (In/Out/Missed, 600 entries)
- Blocked List/Allowed List
- Screen saver
- Voice Message Waiting Indication (VMWI)
- Programmable DSS/Soft keys
- Network Time Synchronization •
- Support Poly wireless headset (Through Poly APD-80 EHS Cable)
- Support Jabra wireless headset (Through Fanvil EHS20 EHS Cable)
- Support Recording (Through Server)
- Action URL / Active URI
- XML Browser
- Customize Background
- uaCSTA
- Customize Incoming Call Ringtone
- Time Plan (timed reboot/timed ٠ forward/timed upgrade/timed config)
- Hidden DTMF
- **Ring From Headset** •
- Customize DSS LED lights •
- Customize BLE
- Customize UI Display

Call Functions

- Call out / Answer / Reject
- Mute / Unmute (Microphone)
- ٠ Call Hold / Resume
- Call Waiting •
- Intercom
- Caller ID Display
- Speed Dial •
- Anonymous Call (Hide Caller ID/Ban • Anonymous call)
- Call Forward (Always/Busy/No Answer) •
- Call Transfer (Attended/Unattended)
- Call Parking/Pick-up (Depending on • server)
- Redial
- Call Back •
- Predial
- Do-Not-Disturb(DND)/Timed DND

Tel: +86-755-2640-2199

- . Auto-Answering
- . SIP SMS
- Voice Message (On server) .
- 6-way Audio Conference .
- Hot Line
- Hot Desking

Direct IP call without SIP proxy

FTP/TFTP/HTTP/HTTPS/DHCP OPT66/SIP

PNP/TR-069

Screenshot

• Syslog

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Web Management Portal

Web-based Packet Dump

Phonebook Import/Export

Firmware Upgrade

Network Firewall

Online Upgrade

➤ 4 Soft-keys

4 Navigation keys

(Microphone))

Handset (RJ9) x1

Desktop Stand x1

Network)

Security Slot x1

Power adapter:

0.84~1.43W

Installation:

Color: Black

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(10 PCS)

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Qty/CTN:10pcs

Device Dimensions:

Desktop Stand

Output:5V/0.6A

1 OK key

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Website: www.fanvil.com

Physical Specifications

Keypad: 36 keys, including

• LCD: 2.4-inch (320x240) Color Screen

> 9 Function keys (Hold/Transfer/Voice

message/Conference/Phonebook

3 Line keys (With LED Indicator Light)

12 Standard Phone Digits keys

Standard RJ9 Handset Wire x1

1.5M CAT.5E Ethernet Cable x1

Status Indicator Light x1 (Red)

standard(only X303P)

RJ9 Port x2: Handset x1, Headset x1

RJ45 Port x2: Network x1, PC x1 (Bridged to

Power: DC5V / 0.6A or PoE(only X303P)

POE complies with the IEEE 802.3af

Input:100-240VAC~50/60Hz

Power Consumption (PoE): 1.32~2.09W

Power Consumption (Adapter):

Working Temperature: 0~45 ℃

Working Humidity: 10~95%

> Wall mountable (optional)

N.W/CTN: 8.95kg(X303), 8.5kg(X303P)

G.W/CTN: 9.66kg(X303), 9.21kg(X303P)

Desktop Stand (45°): 169x201.1x177.4

Wall mountable: 169x120.1x213.7 mm

Outer CTN Dimensions: 460x300x270mm

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Gift Box Dimensions: 250x220x55mm

Desktop Stand (50°): 169x191.9x185.8

/MWI/Headset/Redial/Hands-free)

3 Volume Control keys (Up/Down/Mute

Configuration Export / Import

- Fuzzy Search Call
- Action Plan
- **Dial Plan**
- Call Completion/Auto Redial
- Primary and Backup SIP Servers .
- Default Ext Line
- Multicast

Audio

- HD Voice: HD handset, HD speaker •
- Wideband ADC/DAC 16KHz Sampling •
- Narrowband Codec: G.711a/u, G.723.1, G.726, G.729, G.729A, G.729B, G.729AB, ilbC
- Wideband Codec: G.722. Opus
- Full-duplex Acoustic Echo Canceller (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: 10/100 Mbps Ethernet, dual bridged port for PC bypass
- IP Mode: IPv4/IPv6/IPv4&IPv6 •
- IP Configuration: Static IP / DHCP / PPPoE
- Network Access Control: 802.1x •
- VPN: L2TP / OpenVPN
- VLAN
- LLDP
- 005
- RTCP-XR (RFC3611), VQ-RTCPXR (RFC6035)
- DNS-SRV (RFC 3263) •
- Support Redundant Servers
- Agent
- **Https Mutual Authentication**

Protocols

- SIP v1 (RFC2543) v2(RFC3261)
- UDP/TCP/TLS
- RTP/RTCP/SRTP(Optional/Compulsory)
- STUN
- DHCP
- IPv6
- LLDP
- PPPoE
- 802.1x
- 802.1p/Q •
- L2TP
- OpenVPN •

TR-069

Email: sales@fanvil.com

HTTP/HTTPS

AES128 & AES256

Auto-Provisioning via

Deployment & Management

SNTP • FTP/TFTP

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