

Making every conversation count

NEOX N303/N303P Entry Level IP Phone



NEOX N303/N303P is a SIP phone designed for enterprise communication, offering HD audio quality for an enhanced user experience. It features a 2.4-inch color display with a 320x240 pixel resolution, making it user-friendly and easy to navigate.

This model supports 4 SIP lines and allows for local 6-party conferencing, adding flexibility to communication. Additionally, it is equipped with dual 10/100 Mbps network ports and, for the N303P version, integrated PoE support, ensuring cost-effective and high-performance communication.

Key Specifications

- 2.4-inch color display with a 320x240 pixel resolution
- Supports 4 SIP lines for streamlined multi-line communication
- HD audio on both speakerphone and handset, supporting G.722 and Opus audio codecs
- Supports a **6-party local conference** for effective group communication
- Compatible with EHS wireless headsets for added convenience
- Dual 10/100 Mbps network ports, with PoE integration (available only on N303P) and an optional external power source
- Flexible installation options: desktop stand or wall-mountable designModel

Phone Features

- Local Phonebook: Up to 1000 entries
- Remote Phonebook: XML/LDAP support (1000 entries)
- Call Logs: 600 entries (Incoming, Outgoing, Missed)
- Blocked and Allowed Lists for call management
- Voice Message Waiting Indication (VMWI)
- Programmable DSS/Soft Keys for quick access
- Bluetooth Headset and Wi Fi Support via dongle
- Call Recording: Server-based support
- Screen Saver for display customization
- Network Time Synchronization
- Action URL/Active URI for automation
- uaCSTA for advanced call control

Network Features

- IP Configuration: DHCP, Static IP, PPPoE
- IP Modes: IPv4, IPv6, Dual-Stack (IPv4 & IPv6)
- Dual 10/100 Mbps Ethernet Ports with PC bypass support
- Network Access Control: 802.1X Authentication
- VPN Support: OpenVPN, L2TP

- VLAN Support for network segmentation
- QoS (Quality of Service) for optimized performance
- LLDP and CDP for network discovery and management
- Call Quality Monitoring: RTCP-XR (RFC3611) and VQ-RTCPXR (RFC6035)



Provisioning and Control

- Web Management Portal for easy configuration
- Configuration Management: Export/Import options
- Phonebook Management: Import/Export support
- Firmware Upgrade for seamless updates
- Web-Based Packet Dump for troubleshooting
- Syslog Support for system logging and diagnostics
- Auto-Provisioning: FTP, TFTP, HTTP, HTTPS, DHCP, OPT66, SIP PNP, TR-069

Calling Functions

- · Make, Answer, or Reject Calls
- Hold and Resume Calls
- Mute/Unmute Microphone
- Call Waiting for incoming call alerts
- Redial Last Number
- Speed Dial for quick access to contacts
- Display or Hide Caller ID (Anonymous Call)
- Call Forwarding Options: Always, Busy, No Answer
- Call Transfer: Attended or Blind
- Call Pickup/Parking (Server-dependent)
- Auto-Answer for hands-free convenience
- Do Not Disturb (DND) mode for uninterrupted time
- 6-Way Audio Conferencing for group discussions
- Intercom Function for internal communication
- Hotline Setup for instant calling
- Hot Desking Support for shared workspace environments
- Voice Messaging (Stored on Server)

Network Security and Protocols

- SIP 2.0 over UDP, TCP, and TLS
- Secure Media: SRTP, RTP/RTCP
- VPN Support: OpenVPN, L2TP
- IP Configuration: DHCP, PPPoE

- IPv6 Compatibility for modern networks
- Network Discovery: LLDP (Link Layer Discovery Protocol)
- Authentication: 802.1X Network Access Control
- Encryption: AES-128 and AES-256 for data security
- Firewall Traversal: STUN
- Time Synchronization: SNTP
- File Transfer: FTP/TFTP
- Web Access: HTTP/HTTPS
- Remote Management: TR-069

Audio Features

- Wideband ADC/DAC with 16KHz sampling for superior audio performance
- Full-Duplex Acoustic Echo Cancellation (AEC) for crystal-clear calls
- Wideband Audio Codecs: G.722, Opus
- Narrowband Audio Codecs: G.711a/u, G.723.1, G.726, G.729A/B, iLBC
- Packet Loss Compensation (PLC)
- Adaptive Jitter Buffer (Dynamic)
- DTMF Support: In-Band and Out-of-Band (RFC2833/SIP INFO)
- VQM (Voice Quality Monitoring)
- Advanced Voice Processing:
 - o Noise Reduction (NR)
 - o Automatic Gain Control (AGC)
 - o Voice Activity Detection (VAD)
 - o Comfort Noise Generation (CNG)
 - o Background Noise Estimation (BNE)
- Voice Quality Enhancements:
 - o Packet Loss Concealment (PLC)
 - o Dynamic Adaptive Jitter Buffer
 - o VQM (Voice Quality Monitoring)
- DTMF Support: In-band, Out-of-Band (RFC2833/SIP INFO)
- Handset with Hearing-Aid Compatibility (HAC)
- Handset with HD Voice Microphone & Speaker
 (Frequency Range: 0 ~ 7KHz, for Handset/Hands-free
 modes)



Device Specifications

Display and Controls

• Main LCD: 2.4-inch color screen (320x240 resolution)

• Keypad (36 keys):

o Standard Digits: 12 keys

o Line Keys: 3 (with LED indicator)

o Soft Keys: 4

o **Function Keys:** 9 keys (Hold, Transfer, Voice message, Conference, Phonebook, MWI, Headset, Redial, Hands-free)

o Navigation Keys: 4

o Volume Control: 3 keys (Up, Down,

Mute Microphone)

o **OK Key:** 1

Connectivity and Ports

RJ9 Ports:

o Handset x1

o Headset x1

• RJ45 Ports:

o Network x1

o PC x1 (Bridged to Network)

• DC Power Input: DC 5V / 2A

· Security Slot: 1

Accessories and Installation

• HD Handset (RJ9) x1

Standard RJ9 Handset Wire x1
 1.5M CAT.5E Ethernet Cable x1

• Installation:

Desktop Stand:

• Wall Mountable (Optional):

Environmental Conditions

Operating Temperature: 0°C ~ 45°C
Operating Humidity: 10% ~ 95%

Indicators

• Status Light: Red

Why STL-Neox? -

Trusted Globally: CCaaS/UCaaS solution with 500+ customers worldwid

Certified Excellence: Make in India, Gartner-reviewed, ISO 27001/14001, CMMi Level 5, DOT-TRAI tested

All-in-One Platform: Unifies all communication needs on a single platform

Open SIP Protocol: Avoids vendor lock-in

In-House Development & Customization: Tailored to meet specific client needs **24x7 Professional Support:** Strong post-sales service for seamless operations

Value for Govt. Helplines, Telcos, Data Centers, and Enterprises Multi-Vendor Compatibility: Interoperable with various systems Compliance-Ready: Meets data privacy and security standards

About Sterlite Technologies Ltd. (STL)

STL is a leading global optical and digital solutions company providing advanced offerings to build 5G, Rural, FTTx, Enterprise, and Data Centre networks. The company, driven by its purpose of 'Transforming Billions of Lives by Connecting the World', designs and manufactures in 4 continents with customers in more than 100 countries. Telecom operators, cloud companies, citizen networks, and large enterprises recognize and rely on STL for advanced capabilities in Optical Connectivity, Global Services, and Digital and Technology solutions to build ubiquitous and future-ready digital networks. STL's business goals are driven by customer-centricity, R&D and sustainability. Championing sustainable manufacturing, the company has committed to achieving Net Zero emissions by 2030.



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