

Sangoma P325 Mid-Range IP Phone



Product Name: Sangoma P325 Mid-Range IP Phone

Manufacturer: Sangoma

Model Number: 1TELP325LF

Sangoma P325 Mid-Range IP Phone (1TELP325LF)

The Sangoma P325 stands out with its large 4.3-inch 480x272 color IPS display, enhancing user experience with vibrant visuals. Designed for high productivity, this model integrates essential business applications seamlessly. Its 6-line programmable keys support VoIP accounts and can function as BLF with speed dial, allowing for direct dialing and extension monitoring. Additionally, the P325 includes an innovative paging key, offering 20 pages of virtual keys to expand its functionality even further. This powerful Sangoma device is ideal for businesses looking to optimise communication efficiency.

Sangoma P325 Key Features

- 6-line keys
- Paging key for 20 pages of virtual keys
- Full-color display screen
- High definition audio
- Customizable Busy Lamp Field (BLF) Keys
- Power Over Ethernet (POE)
- Electronic Hookswitch capable headsets (i.e. Sangoma H10 and H20)
- Lock ports to ensure physical security
- Real-time status
- Contacts
- Transfer & conference calls
- Zero Touch Provisioning for all of
- Sangoma's UCaaS platforms

Highly Integrated Productivity Applications with P-Series Phones

Visual Voicemail: See who left you voicemails, along with timestamps and full control to skip and delete.

Queues: Get a full overview of call queue activity; perfect for managers who need to see at-a-glance information about the queues on the system.

Parking: See all parked calls on the system in a visual display and pick up the one you want!

Call Log: In addition to standard call logs, the Call Log app allows you to see the last 50 calls made on the system, persistent across reboots!

Status (Presence): See and change your status and control how calls are handled based on the status. Status options, such as "Away" or "Away - At Lunch" are controlled by the Presence State Module in your phone system GUI.

Contacts: Search for contacts/users or groups, created by the Contact Manager module, to call users easily.

Recording: On-demand recording allows the user to start and stop recording at any time by using a phone button.

Hot Desking (Login/Logout): Log in or out of any phone and have all your extension and user settings follow you.

Conference: See all your conference rooms and view the callers in each. You have full control to admit or deny access, mute callers, or remove callers from a conference room.

Sangoma P325 - Technical Specifications
User Interface

Sangoma P325 Mid-Range IP Phone

- 4.3 inch, 480 x 272 pixel backlit color IPS LCD display
- Line appearance keys: 6 keys
- Rapid dial page scroll key for virtual line keys: Yes
- 6 feature keys
- 4 context-sensitive soft keys
- 6-key navigation controller
- Backlit headset, speaker, and mute keys
- 3-color LED message waiting indicator
- Volume control

Connections

- 4P4C (RJ-9) handset/headset jack
- 2 switched 10/100/1000 Mbps Ethernet ports
- USB-A 2.0 ports: 1
- 12 volt 1.6A DC power port
- 2 Lock Ports
- Electronic Hookswitch signaling port

Phone Features

- Auto-answer
- Headset-answer
- Headset-ring
- Distinctive ring
- Custom ringtones
- Call waiting
- Call transfer (blind & attended)
- Call forward
- Call hold
- Call parking
- Call pickup
- Intercom/Paging
- Message Waiting Indication
- Busy Lamp Field (BLF)
- 3-way conferencing
- Do Not Disturb (DND)
- Redial
- Call timer
- Caller ID display
- Hot Desking Supported on Switchvox 7.8 and above

Phone Applications

- Visual Voicemail
- Contacts
- Presence/Status
- Call Log
- Parked Calls
- Info
- Forward Calls

Dimensions/Weight

Sangoma P325 Mid-Range IP Phone

• 12.3" x 9.0" x 3.5", 3.0 lbs.

Audio Features

- Handset, handsfree, and headset modes
- Full-duplex HD speakerphone with echo cancellation
- Hearing aid compatible
- Multicast audio playback
- Wideband: G.722
- Narrowband: G.711 ulaw/alaw, G.726, G.729a

Network Support

- Static or DHCP IP address assignment
- NTP time synchronization
- Priority queuing
- VLAN, LLDP, NAT traversal, QoS
- 802.1X pass-through
- 802.1X authentication

SIP/VoIP Support

- SIP v2
- TLS/SRTP
- DTMF, RFC4733
- SIP peer-to-peer
- SIP presence (SUBSCRIBE/NOTIFY)
- Per-account digit map/dial plan
- Dial using SIP URI
- Dial using SIP server
- Secure SIP authentication

Management

- Firmware update via browser
- Factory restore via UI
- Remote logging via syslog

Software Minimum Requirements

- Switchvox >=7.9
- FreePBX & PBXact >=15 with Endpoint Manager (latest)

Price: £95.20