

snom 720 IP phone

Pure functionality



- 4-line B/W display
- 18 LED function keys
- 12 identities
- Wideband audio
- Hands-free operation
- Power over Ethernet (PoE)
- WLAN / Bluetooth Headset ready
- VLAN
- 2 x Gigabit LAN ports
and much more

The **snom 720** phone addresses office users that require excellent audio and a large number of programmable PBX-style keys. It combines a state-of-the-art hardware with the proven snom SIP software.

The new audio subsystem was designed for excellent performance in office environments. It supports a large range of codecs, wideband audio, handset and hands-free operation according to TIA-920 standard. The specially designed grips which hold and release the receiver compliment overall user-friendliness by eliminating the noise normally associated with lifting and replacing the receiver.

For network connectivity, the **snom 720** comes with two Gigabit Ethernet ports, one of them suitable for Power over Ethernet. The Ethernet ports are connected to an internal switch that is capable of VLAN tagging, so that an external PC can reside in a different VLAN than the phone. A USB type A connector can be used to expand the functionality of the snom 720. An optional USB stick

can be used to connect the phone over Wireless LAN or enable Bluetooth headsets.

The 4-line B/W display displays caller-ID and other call-related information. Its fully graphic capability makes it possible to show international characters and render content generated from the built-in mini-browser.

Three rows of general-purpose keys with 18 two-color LEDs can be used to implement functions known from traditional PBX systems. Examples include speed dial, busy lamp field (BLF), shared line appearances (SLA), redirection, door opening, and many more features depending on the PBX system used.

The new foot stand allows two positions for the device (33° and 55° angle). It can be setup and changed easily without tools. The **snom 720** comes in a dark grey color that fits the typical office dress code and makes usage traces less obvious than lighter colors.

Like all snom desktop phones, the **snom 720** comes with software that has been used in many installations over the last decade. Configuration can be done manually through the built-in web server or through automatic provisioning. There are extensive management and reporting tools available, such as RTCP-XR or PCAP. Customization can be done in various ways, such as the snom mini-browser protocol.

The phones come with a preinstalled certificate signed by the snom certificate authority which makes it possible to securely provision the phone without manual interaction. In addition to the traditional provisioning mechanisms, it comes with TR 069 for easy deployment in carrier environments.

All in all, the **snom 720** raises the bar for VoIP phones in its class both in terms of voice quality, available features and day-to-day usability.

Technical Data snom 720

GENERAL INFORMATION

- **Weight:** approx. 900g (670g without foot stand)
- **Certifications:** FCC Class B, CE Mark
- **Safety:** IEC 60950-1:2007
- **Power consumption:** 2.8–4.3 W
- **Power over Ethernet/PoE:**
IEEE 802.3af, Class 2
- **Power:** 5 V DC (not included)
- **Ethernet:** 2 x IEEE 802.3/1 Gigabit switch
- **1 x LAN, 1 x PC:** RJ 45
- **Handset:** RJ-4P4C connector
- **Headset:** RJ-4P4C connector or via USB
- **USB ports:** Type A Hi/Full/Low Speed host interface
- **Color:** Anthracite Grey

USER INTERFACE

- Graphical four-line display
- 128 x 48 pixels
- 18 freely programmable keys 4 variable function keys
- Comfortable and intuitive menu structure
- Call indication with LED
- Caller identification
- LED indication for missed and received calls and messages
- Address book with 1000 contact entries
- Speed dialing
- Lists of missed, received and dialed calls (100 entries each)
- Call waiting indication
- Clock with automatic daylight-saving time
- Call blocking (deny list)
- Blocking of anonymous calls
- Handling of up to 12 calls simultaneously
- Auto answer
- URL dialing

- Selection of ring tones, VIP ring tones, integration of customized ring tones
- National Language Support
- DND mode (do not disturb)
- Hands-free talking
- Keyboard lock

ADDITIONAL FUNCTIONS

- Easy WLAN binding via USB stick
- Hold
- Blind and attended transfer
- Call forwarding
- 5-way conference on the phone
- Call park, pickup
- Call completion (Busy/Unreachable)
- Client Matter Code (CMC)
- Automatic Redial
- Group call announcements
- Multiple audio device support
- Unified Communications ready

WEB SERVER

- Built in HTTP, HTTPS server
- Remote configuration/provisioning
- Dialing via Web browser
- Password protection
- Diagnostics (traces, logging, syslog)

SECURITY, QUALITY OF SERVICE

- HTTPS-Server/-Client
- Transport Layer Security (TLS)
- SRTP (RFC3711), SIPs, RTCP
- VLAN (IEEE 802.1X)
- LLDP-MED, RTCP-XR

CODECS, AUDIO

- Wideband Audio
- G.711 A-law, μ -law
- G.722, G.726, G.729AB
- GSM 6.10 (full rate)
- Comfort Noise Generator (CNG)
- Voice Activity Detection (VAD)

SIP

- RFC3261 compliant
- UDP, TCP and TLS
- Digest authentication
- PRACK (RFC3262)
- Error code indication
- Reliability of provisional responses (RFC3262)
- Early media support
- DNS SRV (RFC3263), redundant server support
- Offer/Answer (RFC3264)
- Message Waiting Indication (RFC3842), Subscription for MWI events (RFC3265)
- Dialog-state monitoring (RFC 4235)
- DTMF
- STUN client (NAT traversal)
- NAPTR (RFC 2915), rport (RFC 3581), REFER (RFC 3515)
- Bridged line appearance (BLA)
- Auto provisioning with PnP
- Presence/Buddy-list feature
- Busy lamp field support (BLF)
- Presence publishing

INSTALLATION

- Automatic software updates
- Automatic loading of settings via HTTP/HTTPS/TFTP
- Completely configurable through Web interface
- Static IP, DHCP support
- NTP