

VOIP PHONE SNOM 300

SIP based VoIP Phone



- Two-line display (2x16 characters)
- 27 keys, 7 LEDs
- 6 programmable function keys
- 2x IEEE 802.3 10/100 Mbps switch
- 4 identities
- Headset connection
- SIP RFC3261
- Security: SIPS/SRTP, TLS
- STUN, ENUM, NAT
- Codecs: G.711, G.729A, G.723.1, G.722, G.726, GSM 6.10 (full rate)



Usability



Security



Interoperability

The IP telephone with the latest VoIP technology for the discerning caller! Ideal for ITSPs, offices, and private users.

As the basic model of the snom business telephone family, the SIP telephone snom 300 fulfils the requirements of VoIP telephony and additionally offers numerous functions that are indispensable for everyday working life. Due to its excellent cost-performance ratio, it is ideal both for ITSP packages/products and for small and medium-sized enterprises, home offices, and private users.

The snom 300 sets new standards when it comes to user-friendliness: A two-line LCD display shows call information, and the menu-driven user interface provides simple feature management.

The snom 300 is synonymous with effectiveness, time saving and flexibility: It provides all necessary office functionalities, for example choice of outgoing identity, "busy" indication, or picking up calls. Four identities are possible.

Six user- or administrator-configurable (or carrier-preconfigurable) function keys can easily be allocated to arbitrary menu functions like DND, redirection, deny, etc., or assigned to multiple lines.

The snom 300 comes factory-equipped with two of its six programmable keys configured as line appearances. You can configure up to all six function keys this way – flexible enough to suit the needs of every user.

Are you afraid of unsolicited eavesdroppers, data theft, or spam calls? Thanks to the snom 300 you have no need to worry. Due to the security standards SRTP and SIPS, the snom 300 is just as secure as the business models snom 360 and 320. This makes it perfectly suited for use in sensitive environments such as in banks and medical or military establishments.



TECHNICAL DATA SNOM 300

GENERAL FEATURES

• Dimensions:

approx. 20x18.5x12 cm

• Weight: approx. 730 g

Safety: IEC 60950-1:2001,
CB Test Certificate: DE 02011039

Certifications: FCC Class B,

CE Mark

• **Power consumption:** 1.9–2.1 watts

CONNECTORS

• 1x LAN, 1x PC: RJ45 (Ethernet)

• Power: 5 V DC

• Ethernet: 2x IEEE 802.3 10/100 Mbps switch

• Power over Ethernet: IEEE

802.3af, Class 1

Handset: RJ11 connectorHeadset: RJ11 connector

USER INTERFACE

- 2x16 character display, backlit
- 27 keys, 6 programmable function keys with LEDs
- Caller-ID
- Message waiting indication LED
- Address book (100 entries)
- · Address book import/export
- · Speed dialing
- Local dial plan
- · Number guessing
- Lists of missed, received and dialed calls (100 entries each)
- · Call waiting indication
- Clock, daylight saving, call-timer
- Call blocking (deny list)
- Blocking of anonymous calls
- · Handling of 4 simultaneous calls
- Menu-driven user interface
- Ring tone selection, import of individual ring tones
- URL Dialing support
- National language support for selected languages (NLS)

- · Do not disturb
- Speakerphone (full duplex)
- Auto answer mode
- · Keyboard lock

CALL FEATURES

- Hold
- Blind transfer, attended transfer
- Music on hold support (only via PBX)
- Divert
- Conferencing (3-way conference bridge on phone)
- Call park, call pick-up (only via PBX)
- Call completion
- Client Matter Code (CMC)
- Call waiting/switching between calls
- Redialing
- · RTP multicast paging
- Multiple audio device support

WEB SERVER

- Embedded web server HTTP/HTTPS
- Easy configuration of the phone, remote configuration
- Dial from web interface
- · Password protection
- Diagnostics (tracing, logging, syslog)

SECURITY, QUALITY OF SERVICE

- HTTPS (server/client)
- Transport Layer Security (TLS)
- SRTP (RFC3711), SIPS
- RTCP
- VLAN (802.1 pq)

CODECS, AUDIO

- G.711 A-law, μ-law
- G.729A, G.723.1, G.727
- GSM 6.10 (full rate)
- G.722
- Comfort noise, voice activity detection

SIP

- RFC3261 compliance
- UDP, TCP and TLS
- Digest/basic authentication
- Loose routing and strict routing support
- PRACK (RFC3262)
- Error-information support
- 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)
- In-band DTMF/out-of-band DTMF/ SIP INFO DTMF
- STUN client (NAT traversal)
- ENUM (RFC3261), NAPTR (RFC2915), rport (RFC3581), REFER (RFC3515)
- Event list subscription support (RFC4662)
- Bridged line appearance (BLA)
- Auto provisioning with PnP
- Busy lamp field support (BLF)

INSTALLATION

- Automatic software update
- Automatic settings retrieval via HTTP/HTTPS/TFTP
- Installation via web interface
- Static IP provisioning, DHCP
- NTP

For more information, please contact your snom partner:

Version 1.2/15 Nov 2007

Copyright © 2000-2007 snom technology AG. All rights reserved. snom is a registered trademark of snom technology AG and affiliates in Germany, USA and certain other countries and regions. Unless specified otherwise,

all trademarks, in particular product names, in this document are legally protected trademarks of snom technology AG. Other trademarks or registered trademarks mentioned in this document are the property of their respective manufacturers or owners. Product specifications contained in this document are subject to change without notice.

snom technology AG

Gradestraße 46 D-12347 Berlin tel/enum: +49 30 39833-0 fax: +49 30 39833-111 sip: info@snom.com mail: info@snom.com

www.snom.com