



SwyxPhone S315

Leading edge SIP phone in a classic design

The new SwyxPhone S315 is an economical entry-level SIP phone that enables enterprises to take full advantage of the benefits of VoIP telephony. The SwyxPhone S315 combines familiar design elements

with modern technology providing an excellent cost-performance ratio. By supporting the standard SIP protocol, the telephone provides future-proofing and can be integrated flexibly into various IT infrastructures.



Increased efficiency through extensive business telephony functions

The SwyxPhone S315 delivers as standard, all capabilities that are necessary for daily business activities, such as call hold, call transfer, call swap, call lists and 3-party-conferencing.

In addition to these frequently used functions, the user has predefined function keys and programmable speed dials (name keys) that can be configured for the individual's requirements.

The speed dial keys provide the additional advantage of showing the presence status of the person to whom the speed dial is allocated.

Additional functionality through integration with SwyxWare

The SwyxPhone S315 is part of the innovative SwyxWare product family and is optimised for smooth interaction with Swyx components. Users therefore benefit further from proven capabilities which enable more efficient working practices on a daily basis. For example, Group Calls allow a group of customer support co-workers to be provided with its own direct dial number so

that any member of the group can pick up an incoming call on the SwyxPhone S315. In addition, the 'Notification/Pick-Up Calls' feature enables a user to pick up a call originally destined for another (absent) colleague, so that the caller - for example, a valued customer - may have his enquiry answered more quickly.

The Dialler Mode allows the control of a SwyxPhone S315 via the easy to use graphical interface of the SwyxIt! soft-phone so that you can always use the best tool. Nearly all SwyxIt! functions are available via the Dialler Mode e.g. dialling from the phone-book, from the call lists, or from other applications.

The SwyxPhone S315 offers the following key features:

- Support for all important telephony functions: call hold, call swap, re-dial, local tele-conference, etc.
- Additional functions for more efficient working practices: group calls, notification/pick-up calls, intercom
- Telephone book with up to 100 entries
- Call lists (dialled calls, missed calls and received calls)
- Easy to read, 2 line display
- Function keys and programmable speed dials (7 of each)
- Speed dials showing presence status
- Handsfree-speaking, connections for microphone and headphone
- Power over Ethernet (PoE)

Simple start-up and maintenance

The SwyxPhone S315 is very easy to implement and maintain: a set-up wizard within the telephone ensures a fast installation - a 'plug and play' solution. SwyxServers that are already present within the network are

automatically detected and existing configurations will be adopted (auto provisioning).

Furthermore, all telephony and network settings can be easily

processed within the web-interface of the SwyxPhone S315 which is provided by the in-built web-server. Software updates can also be applied in this same simple manner by the administrator or authorised user.

SwyxPhone S315 – Main Features

Telephony functions

- Call hold, call swap, call forwarding, call pick up, group calls, do not disturb, local 3 party conferencing, re-dial, voicemail enquiry, Intercom
- Local call lists (missed calls, received calls, re-dial list)
- LCD-based telephone menu
- Personal phone book with up to 100 entries
- Handsfree-speaking, connections for microphone and headphone
- 7 predefined function keys and 7 programmable speed dials (name keys)
- Speed dials display presence status
- 2 additional status LEDs (for missed calls and for mute)
- Message Waiting Indicator (MWI) for voicemails
- Large handsfree button
- Call forwarding options
- Integration with SwyxIt! soft-phone via SwyxIt! Dialler for SwyxPhone S315

VoIP

- Support for standard SIP protocols e.g. RFC 3265, RFC 3261, RFC 2327, RFC 2976 (DTMF Info Method), RFC 3711
- Codecs: G.711, G.729
- STUN support

Network

- 10/100 Ethernet Switch, DHCP, Fixed IP
- Name resolution : DNS, SRV Records
- NTP support

Configuration & Administration

- Start-up via wizard
- Configuration via telephone and web-interface
- Firmware update via web-interface or TFTP

Scope of delivery

- Quick start manual, Ethernet cable (3m), power pack, CE conformity explanation

