



## IPAC 101-2 S

VoIP Door Entry Phone **for audio transmission**, « hands-free », vandal-resistant:

- **Pictograms display and automatic speech announcements to meet the requirements for accessibility for people with disabilities**

- 1 direct call button

The IPAC 101-2 S allows for:

- a point-to-point communication (Peer to Peer) **or**
- the connection via a SIP server.

### Functions

- Telephone
- Full duplex voice communication

### Technical data

- 1 call button
- Redial if busy or if no answer (1 - 4 call numbers)
- Management of call parameters: communication time, button activation time, ring time for outgoing calls, volume...
- Management of time lock zones
- Pictograms display associated with product functions
- Automatic speech announcements (dialling, communication ..., door opening)
- HD audio quality
- Media incryption (audio and video): SRTP/ZRTP/ SIP-TLS
- 1 relay for door open command **or** remote control of external elements (line seizure information)
- 1 input for external contact **or** voltage **with the possibility to define time lock zones**
- Updates:
  - LDAP-update of the phonebook
  - system update by downloadable file
- Real-time monitoring of device status:
  - On access code keying, outgoing calls, door opening, loss of SIP server
  - In case of power failure
- Real-time display of the device screen on the web page

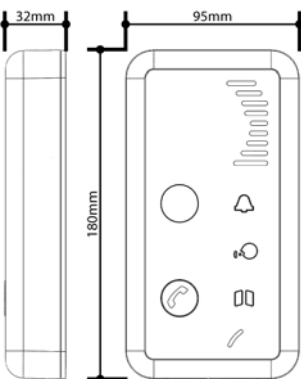
### Power supply

- Network: POE **or**
  - PoE+ injector and **IP network access protection - AMPHITECH NetCut** **or**
  - External power supply unit: 24 VDC - **AMPHITECH BAS 2415**
- Power consumption:
- Idle: 84 mA
  - In communication/Audio adjustment medium level: 170 mA
  - In communication/Audio adjustment maximum level: 270 mA

### Mechanical design

- Surface mount
- Dimensions 180 x 95 x 32 mm
- Degree of protection: IP 55 - IK 08
- Temperature range: -20°C to +60°C
- Stainless steel faceplate 2.5 mm, ZAMAK housing



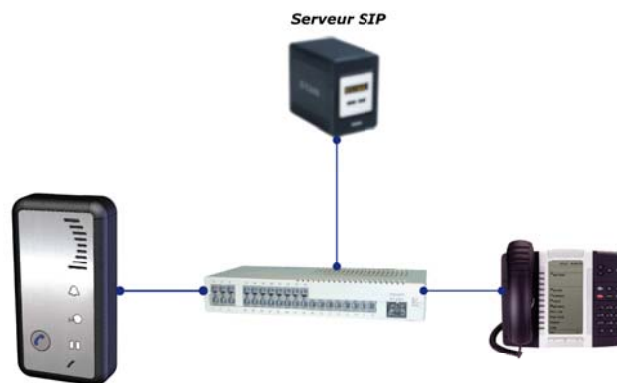


## IPAC 101 -2S

*Point-to-point communication (Peer to Peer)*



*Connection via a SIP server*



### Configuration

- Access via web server
- Basic programming; advanced programming via specific web interface
- Network parameter setting, SIP account, audio- and video codecs
- Point-to-point communication (Peer-to-Peer)
- Communication via SIP server (group call, conference call, queue management, messaging...)
- Management of 3 SIP accounts on different IP-PBX
- Communication protocols: SIP (RFC3261), IPV4, TCP/UDP, HTTP, HTTPS, RTP, DHCP/STATIC/ NAT, RFC 6086 INFO Method, DTMF RFC 2833, RFC 2976 SIP INFO
- Audio codecs: G.722, G.711u, G.711a, GSM, Speex 8k, Speex 16k, Speex 32k, G.726-16, G.726-32, G.726-24, G.726-40, AAL2-G.726-16, AAL2-G.726-32, AAL2-G.726-24, AAL2-G.726-40, opus, AMR.-32
- Report of system events: downloadable files, SYSLOG, notifications by e-mail (SMTP-Client)
- Language selection (configuration, operation mode): French, German, English, Spanish, Portuguese
- Type of time setting: manually or via NTP server
- Automatic clock change – winter time / summer time

