

pbxnsip

"The voice communications platform of choice."

Communications Server CS 410/425

The pbxnsip CS 410/425

The small office appliance from pbxnsip integrates analog telephone lines within the pbxnsip software. It alleviates the need to run the software on a separate PC reducing the power consumption and administrative effort. Using solid-state hardware minimizes the risk associated with hard disks and CPU fans.

It is the perfect solution for small to medium sized businesses (SMB) that want to take advantage of VoIP's benefits and have the option to keep their existing phone lines. For example, local calls can be terminated through the 4 FXO or (PSTN) lines while international calls can be sent through the Internet via your favorite internet service provider (ITSP).

Modern System Design

The CS 410/425 is built on a System on Chip (SoC) design that integrates all essential PBX functions on one chip. This reduces the hardware costs and increases the system stability. Using standard memory like NAND-Flash makes it possible to store large amounts of data in the system. A digital signal processor echo cancellation and gain control on the PSTN side.

Key Features & Benefits

- Built-in 4 Port FXO PSTN Gateway
- 10 Extension Support (CS 425: 25 Extensions) optional
- Paging and Music on Hold audio connectors
- Full PBX features including:
 - Voicemail, Auto Attendant, Intercom, Paging, Conference Server, Hot Desking, Agent Groups
 - Cell Phone Integration, find me/follow me
 - SIP Trunking, NAT support
 - Full version 3.0 feature set

The CS 410/425 comes with two Ethernet ports. This makes it possible to use a private IP address in the LAN and at the same time connect phones in remote locations to the public IP address on the second Ethernet interface (WAN).

Full Feature Set

The system includes all the features from the IC based pbxnsip editions. This includes standard features like voicemail, auto attendant or conferencing, but also advanced features like call barge in or cell phone integration.

Popular CPE PBX Features

The CS 410/425 supports music on hold input from a regular audio jack input. Music on hold can be provided from CD or MP3 player as well as local radio stations. The system has a paging output connector. This makes it possible to use existing overhead paging infrastructure. In the LAN the system can also send RTP multicast traffic and use multicast-enabled devices for office audio paging.

Plug and Play

The CS 410/425 IP PBX supports plug and play of popular IP phones like Polycom, snom, Aastra, and Linksys. The PBX itself can be managed through the web interface. It also supports SNMP for remote quality supervision with standard network management tools.

GOOD LOOKING



COOL DEVICE!

save more than 3000 kWh per year* compared to a PC server.

HARD WORKING



- Plug & Play and Remote Management
- Integrates with Microsoft SIP-compatible products
- Full Security Support (TLS/SRTP/HTTPS)
- HTTP-based management and full OS access
- Now with 1 GB of built-in flash memory
- Low power consumption, fanless operation (less than 7 watts)

* PC with 400 Watt takes 3500 kWh, CS410 takes around 200 kWh per year.

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Technical Specifications for the pbxnsip CS 410/425

Main Features

- SIP compliant
- Built-in 4 Port FXO PSTN Gateway
- 2 x RJ-45 Network Interfaces
- Music In and Paging Out audio connectors
- Small size and fan less / disk less operation
- T.38 (FAX) relay

Accounts

- Auto Attendant
- Extensions
- Paging Groups
- Hunt Groups
- Agent Groups
- Outbound Call Authentication
- IVR Nodes
- Service Flags

Security Features

- Loadable SSL/TLS certificate
- TLS and SRTP support
- HTTPS web interface
- Secure provisioning
- Password and PIN per extension
- SSH access to the system

Mobility Support

- Call forking to cell phone
- Voicemail triggers call to cell phone
- Inbound call cell phone detection
- Camp on from cell phone
- Hot desking support

Call Supervision

- Call barge in
- Training mode
- Listen in

Plug and Play

- TFTP, HTTP and HTTPS support
- One-shot password provisioning
- Time zone provisioning
- PBX provisioning and configuration templates

Trunking

- B2BUA (IP-Gateway) architecture
- Registration and gateway trunks
- ENUM support
- CO-line emulation
- ANI number presentation
- DID routing
- Extension-based dial plans

MoH, Paging and Intercom

- Multiple MoH sources (RTP, File, audio input)
- Multiple audio paging output (audio output, RTP multicast)
- Intercom through star code

Auto Attendant

- Dual language support
- Prerecorded standard destinations
- Day/night mode, holidays
- Dial by name
- Anonymous call intercept
- Black and white list management
- Camp on

Conference Subsystem

- Conference mixer
- Instant conference
- Conference scheduler with Email invitation

Voicemail System

- Private and shared voicemail
- Voicemail notification through email
- Message Waiting Indication (MWI)
- Voicemail commenting
- Support for external voicemail system (e.g. Microsoft Exchange 2007 UM™)

Waiting Queues

- Up to ten announcements
- MoH mixing with announcements
- Agent recovery time
- Call pickup from queue

- Call escalation
- Day/night mode, holidays
- Web based queue status display

Hunt Groups

- Serial and parallel search
- Day/night mode, holidays
- Distinctive ringing
- Custom IVR recordings and routing decisions (IVR Node)

Call Redirection and Treatment

- Anonymous caller ID intercept
- Do not disturb
- Redirect on busy, timeout, always
- Multiple registrations per extension
- Call park, call pickup, call retrieve
- Star code-based transfer
- Last call return, redial
- Caller-ID blocking

Address Book

- Personal/domain level address book
- Address book import
- PNP downloadable Address Book

Presence and Instant Messaging

- Presence agent for presence information
- Instant Messaging support
- Support for dialog state (BLF) System Management

Management & CDR

- Performance monitoring and load protection
- SNMP agent
- CDR export through SOAP interface
- Built-in session border functionality for remote offices

Languages

- Multiple web interface languages
- Multiple audio languages
- Multiple simultaneous time zones



European Office
pbxnsip Europe Ltd
Aspect Court
47 Park Square East Leeds
LS1 2NL United Kingdom

pbxnsip Inc.

100 Burttt Road, Suite 200
Andover, MA 01810, USA
<http://www.pbxnsip.com>

