

SlipperPABX

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Reducing operating costs and boosting productivity are major priorities of business today. Slipper gives your organisation the competitive edge to effectively manage these challenges.

A Slipper phone system is the next generation IP (Internet Protocol) PABX that offers superior functionality and low cost of ownership to the business market. SlipperPABX is a SIP (Session Initiation Protocol) standards-based Voice over IP system, making it easier and affordable for clients to integrate any SIP compliant device into their network. SlipperPABX is an integral part of your entire companies voice solution, branch offices, home offices and roaming users can all use the one Virtual PABX, lowering investment, administration and transmission costs.



Feature List

The features available in a SIP based IP Telephony solution is the end result of the unique combination of the: SlipperPABX, Voice gateway and the particular SIP phones used. As a result, the following list is only *indicative* and not *authoritative* of the full set of functionality delivered in any one particular Slipper implementation.

Please speak to your Slipper consultant for more detail regarding any required feature not mentioned in this list.

Place calls

- Voice
- Video
- Whiteboard sessions

Hold

Call Waiting

Attended Transfer

Blind Transfer

Network topology-aware codec negotiation

Caller-ID

- Numbers
- Text
- Email-address style
- Number to Name database lookup no maximum number of entered numbers

Caller-ID manipulation

- Caller-ID with Suppression [1]
- Dialed-Number Caller-ID [2]

Call Blocking / Call Authorisation

Interactive Voice Response System

Automatic Route Discovery (ENUM)

Clustering (including cross-site Clustering)

Least Cost Routing

- Per User LCR [3]
- Sticky call routing]

Call Forward

- Busy
- No Answer
- Unreachable
- Ad-hoc
- SIP headers (can trace the forwarding path of this call)

Hunt Groups

Call Pickup

- Directed
- Group
- Any

Instant Messaging

Web Interface

- Phone provisioning
- Active Calls (Source and Destination)
- Call Accounting

Voice Mail Service

- Integrates with existing email system via industry-standard SMTP and POP3
- Message Waiting Indicators

Music on Hold

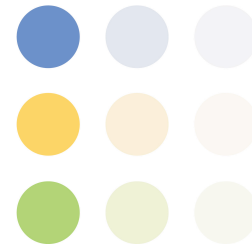
[1] Supports presenting Caller-ID to carrier for billing purposes while marking caller-ID suppressed

[2] Change the text of the Caller-ID to reflect the number dialed, so the Caller-ID text can say "Jim/Jane Doe" so you know jane called Jim's number. Useful when the handset doesn't support the SIP "Diversion" header and won't present it on-screen.

[3] In Per User LCR, users with different privilege levels can have different LCR tables, eg, a manager in Adelaide calling Sydney might do call hop-off in Sydney, but if the Sydney gateway is full or the link is down, it might fall back to an STD call out the local gateway, while a member of the clerical pool may only be able to do call hop-off in Sydney and have to wait if the Sydney gateway is congested.

iagu networks who?

Iagu Networks is a leading Australian IT consulting company specialising in IP Telephony Solutions, Cisco Technologies, Universal Dial and Network Security. We provide services to organisations ranging from Carriers, Internet Service Providers, Banking, and major Agribusiness and Exploration corporations in many countries.



Core Competencies

- IP Telephony Solutions
- Internet Service Provider Consultancy Services
- VPN Configuration and Management
- Firewall Integration and Management
- Internet Security Advice and Audit
- Network Design and Audit Services
- Facilitate Provision of Internet Access
- Universal Dial Access

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