

PGW 2200 Softswitch – CLIP and CLIR Support

Document ID: 53140

Introduction

Prerequisites

Requirements

Components Used

Conventions

Resolution for Presentation Bit Setting in SS7 IAM Message

For the Cisco HSI

For the Cisco PGW 2200

For the Trunk Group Properties

NetPro Discussion Forums – Featured Conversations

Related Information

Introduction

This document explains how to configure calling line identification presentation (CLIP) to calling line identification restriction (CLIR) support when you place a call from the H.323 network to the SS7 network. It applies to the call control PGW 2200 Softswitch version 9.4(1) and higher and to the Cisco HSI solution.

CLIP is a service provided to the called party that allows the display of the calling number (caller ID). The user-provided calling number must be transported from the caller to the called party.

CLIR is a service provided to the calling party that allow the calling party to indicate whether or not the calling number is to be displayed to the called party. It sets a calling number presentation indicator to allowed or restricted. Regulations require that network administrations remove the calling number before it is sent to the called party, if the calling party has so requested.

Prerequisites

Requirements

Readers of this document should have knowledge of the Cisco Media Gateway Controller Software Release 9 Documentation.

Components Used

The information in this document is based on these software and hardware versions:

- Cisco PGW 2200 Software Releases 9.4(1) or higher
- Cisco HSI software 4.1 or higher

	SS7	H.246 Annex c H.450 Support for PSTN Side CLIP or CLIR with H.450.8	H.225
CLIP	Q.731.3	C.6.2.1	Setup/CGN source

			address SI/PR
CLIR	Q.731.4	C.6.2.2	Setup/CGN source address SI/PR

The information in this document was created from the devices in a specific lab environment. All of the devices used in this document started with a cleared (default) configuration. If your network is live, make sure that you understand the potential impact of any command.

Conventions

For more information on document conventions, refer to the Cisco Technical Tips Conventions.

Resolution for Presentation Bit Setting in SS7 IAM Message

The local exchange function establishes the SS7 IAM message in the SS7 network, after a number of basic decisions about the supplied caller ID (if any) are made. If no CLI is supplied or a network CLI is required, then the Cisco PGW 2200 must be able to supply a CLI to the outgoing call on a trunk group basis.

Make the changes in the next section while in call control mode.

For the Cisco HSI

This feature allows the CLI presentation and screening indicators to be passed from ISDN User Part (ISUP) IAM messages to the Q.931 portion of the H.323 setup messages, and vice versa. Issue these commands to configure this on the Cisco HSI:

```
gw mm1> prov-sta:srcver="active",dstver="presentation"
gw mm1> prov-add:name=SYS_CONFIG_STATIC,ClipClirSupported=enabled
gw mm1> prov-add:name=CCPackage,A_CC_AnumDataSI=1
gw mm1> prov-add:name=CCPackage,A_CC_Clr=1
gw mm1> prov-cpy
```

Restart the Cisco HSI. You must restart HSI because these are some of the static variables that must be read at start time or switch-over.

Note: To switch the feature on, the element must be present and given any arbitrary (string) value. To switch the feature off, the element must be given no value (" "). Alternatively, it can be deleted from the configuration.

- On

```
prov-add/ed:name=SYS_CONFIG_STATIC, ClipClirSupported=enabled
```

- Off

```
prov-ed:name=SYS_CONFIG_STATIC, ClipClirSupported=""
```

or

prov-dlt:name=SYS_CONFIG_STATIC, ClipClirSupported

Refer to HSI Data Collection for Technical Support Service Requests.

For the Cisco PGW 2200

If the presentation restriction bit is still set in the SS7 IAM message, also check the information in the routing plan of the Cisco PGW 2200. In the event that you have to add the command to the routing table, check the dw information. If it set to 1, then it will always go out as restricted.

```
numan-add:resulttable:custgrpId="dp01",name="PresEnabled",
resulttype="CG_PRES_IND",dw1="1",setname="cisco1"
```

Note: These are the meanings of the dw values:

- 1 Restricted
- 2 Unavailable
- 3 Allowed

For the Trunk Group Properties

Also check this configuration settings on the Cisco PGW 2200 trunk group basis: time-division multiplexing (TDM), Extended ISDN User Part (EISUP), and session initiation protocol (SIP).

Egress (these apply to messages that exit the Cisco PGW 2200, both backward and forward):

- SendAddressInCgPN If set to 0, this setting unconditionally removes address digits.
 - ◆ 0 Do not include address digits in the Calling Party Number.
 - ◆ 1 Include address digits in the Calling Party Number.
- SuppressCLIdigit If set, this setting removes address parameters, if the presentation indicator is set to restricted.
- CgPNInclude (on SIP only) Removes the Calling Party Number parameter, if the presentation indicator is set to restricted.

Note: Refer to Table A-62 Software Release 9.x Properties in Components and Properties for a Media Gateway Controller.

If the variant is International Telecommunication Union (ITU) Q767_Base on the call incoming side, and if the SS7 IAM message is coming with nature of address (NOA) set to 000011 (not used), then this results in a missing A_number. ITU Q767_Base only supports the sections C.3.7 (Called Party Number) and C.3.8 (Calling Party Number):

```
C.3.7 Called party number:
b) Nature of address indicator
0000000 spare
0000001 , not used
0000010 , not used
0000011 national (significant) number
0000100 international number
```

```
C.3.8 Calling party number
b) Nature of address indicator
0000000 spare
0000001 , not used
0000010 , not used
0000011 , not used
```

0000100 international number

Here you can see that 0000011 is not supported for Calling Party Number.

If you run a Cisco PGW 2200 Message Definition Language (MDL) trace, you will see the error as validation failed for RANGE or MAP:

```
!--- Information from the MDL trace.
!--- Output suppressed.

reading element Q767CallingPartyNumber
      reading field RAW
        56 bits read
      ok
      reading field DATA
        reading field oddEven
          '1'B
        ok
      reading field cgpnoa
        reading field ok
          '0000 011'B
          validation failed for RANGE or MAP
        not found
      reading field err
          '0000 011'B
      ok
```

NetPro Discussion Forums – Featured Conversations

Networking Professionals Connection is a forum for networking professionals to share questions, suggestions, and information about networking solutions, products, and technologies. The featured links are some of the most recent conversations available in this technology.

NetPro Discussion Forums – Featured Conversations for Voice
Service Providers: Voice over IP
Voice & Video: Voice over IP
Voice & Video: IP Telephony
Voice & Video: IP Phone Services for End Users
Voice & Video: Unified Communications
Voice & Video: IP Phone Services for Developers
Voice & Video: General

Related Information

- [Cisco PGW 2200 Softswitch Tech Notes](#)
- [Cisco Signaling Controllers Technical Documentation](#)
- [Voice Technology Support](#)
- [Voice and Unified Communications Product Support](#)
- [Recommended Reading: Troubleshooting Cisco IP Telephony](#)
- [Technical Support – Cisco Systems](#)

[Contacts & Feedback](#) | [Help](#) | [Site Map](#)

© 2008 – 2009 Cisco Systems, Inc. All rights reserved. [Terms & Conditions](#) | [Privacy Statement](#) | [Cookie Policy](#) | [Trademarks of Cisco Systems, Inc.](#)

Updated: Feb 02, 2006

Document ID: 53140
