

How to Configure VOIP Connections with the Skinny (SCCP) Firewall Plugin

<https://campus.barracuda.com/doc/41115962/>

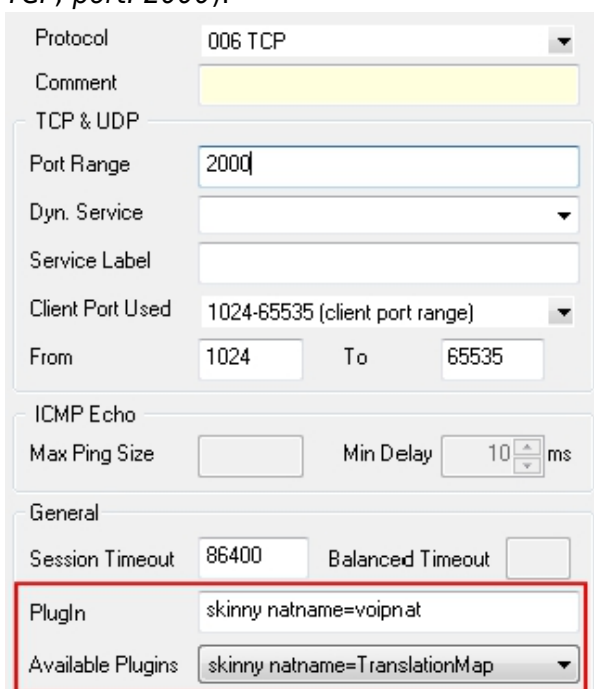
SCCP (Skinny Client Control Protocol) is the protocol used by Cisco callmanager software for VOIP telephony. The VOIP connection is made up out of two separate connections: the control connection handling signaling and RTP data streams for the audio/video transmissions. In order to open the necessary dynamic ports for the RTP connection you need to use the Skinny firewall plugin. The plugin monitors the signaling connection between the VOIP phone and the Cisco callmanager on TCP port 2000. When a new call is initiated the plugin will interpret the packet containing the connection information and open the ports. Similarly these ports are closed when the plugin detects the corresponding call release packet in the skinny control connection.

In this article

Step 1. Create Service Objects for Signalling and Streaming Purpose

For information concerning service objects, see [How to Create Service Objects](#). The skinny plugin has two optional parameters which can be entered in the **Plugin** field:

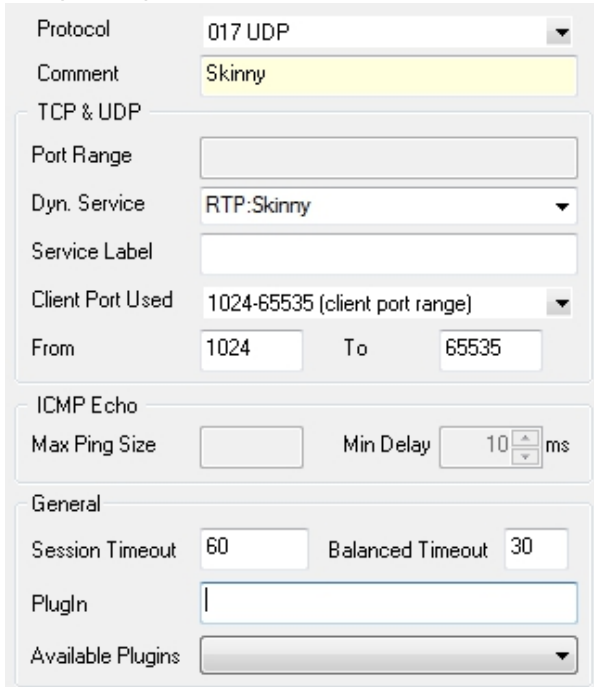
- **natname** - is a reference to a Network Address Translation Map in the **Connections** tab in the firewall rule set (syntax: *skinny natname=<natname>*) and handles the signalling (protocol: *TCP, port: 2000*).



Protocol	006 TCP	
Comment		
TCP & UDP		
Port Range	2000	
Dyn. Service		
Service Label		
Client Port Used	1024-65535 (client port range)	
From	1024	To 65535
ICMP Echo		
Max Ping Size		Min Delay 10 ms
General		
Session Timeout	86400	Balanced Timeout
Plugin	skinny natname=voipnat	
Available Plugins	skinny natname=TranslationMap	

If this option is not specified then the default value *RTP:Skinny* (see below) is used instead. No address translation is performed for the RTP media streams if there is no matching entry in **Connections**.

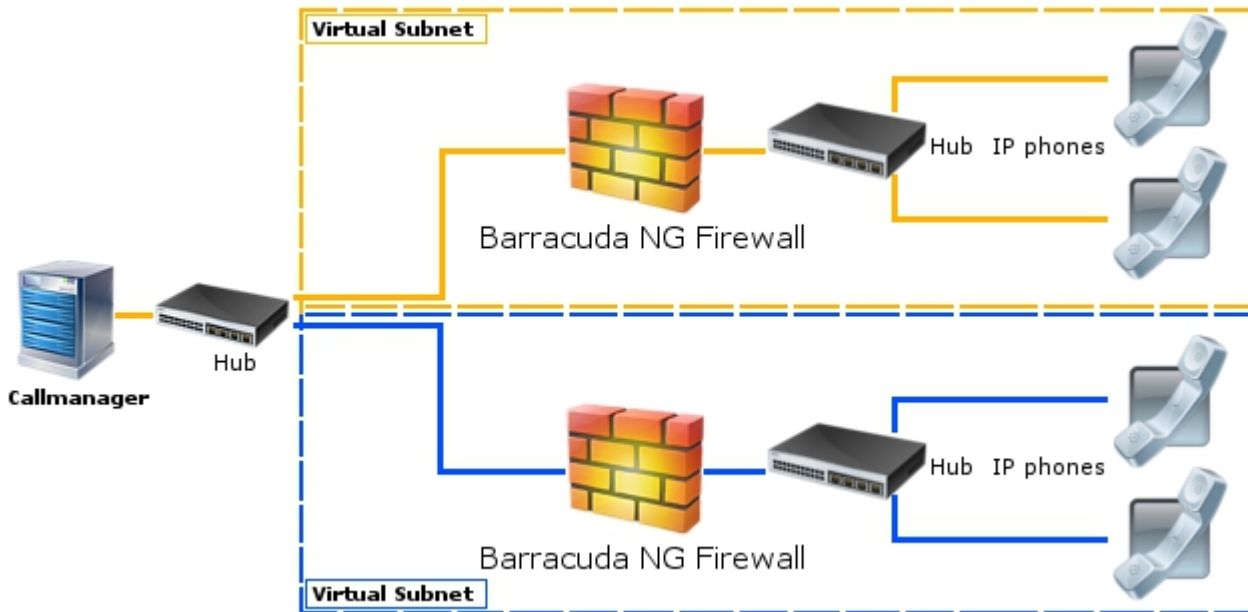
- **srvname** - is a reference to a Dyn. Service label that data fills a service object with the data stream of skinny calls (syntax: *skinny [srvname=<srvname>]*) (protocol: *UDP*). The service object can be referenced by a firewall rule in order to forward the media streams between the call participants. The default value of *srvname* is *RTP:Skinny*.



The screenshot shows the configuration interface for a service object. The 'Protocol' is set to '017 UDP'. The 'Comment' field contains 'Skinny'. Under the 'TCP & UDP' section, 'Port Range' is empty, 'Dyn. Service' is set to 'RTP:Skinny', 'Service Label' is empty, 'Client Port Used' is '1024-65535 (client port range)', and 'From' is '1024' and 'To' is '65535'. Under the 'ICMP Echo' section, 'Max Ping Size' is empty and 'Min Delay' is '10 ms'. Under the 'General' section, 'Session Timeout' is '60' and 'Balanced Timeout' is '30'. The 'PlugIn' field is empty, and 'Available Plugins' is a dropdown menu.

Step 2. Create Translation Map (optional)

If network address translation is done between caller and callee an address translation map has to be defined, translating the real IP address of the participants to virtual addresses that are routeable for all nodes in the VOIP network. For more information, see [How to Create NAT Tables \(Translation Maps\)](#).



The name of the map must match the option of the **natname** parameter of the skinny firewall plugin configured above. The Original Address/Net is the physical IP subnet of a node whereas the Translated Address/Net is the virtual address.

Edit / Create an Address Translation Map ✖

Name: Translation Color

Description:

Connection Timeout: Seconds Same Port

Entry

Original Address/Net/Range:

Translated Address: Proxy ARP

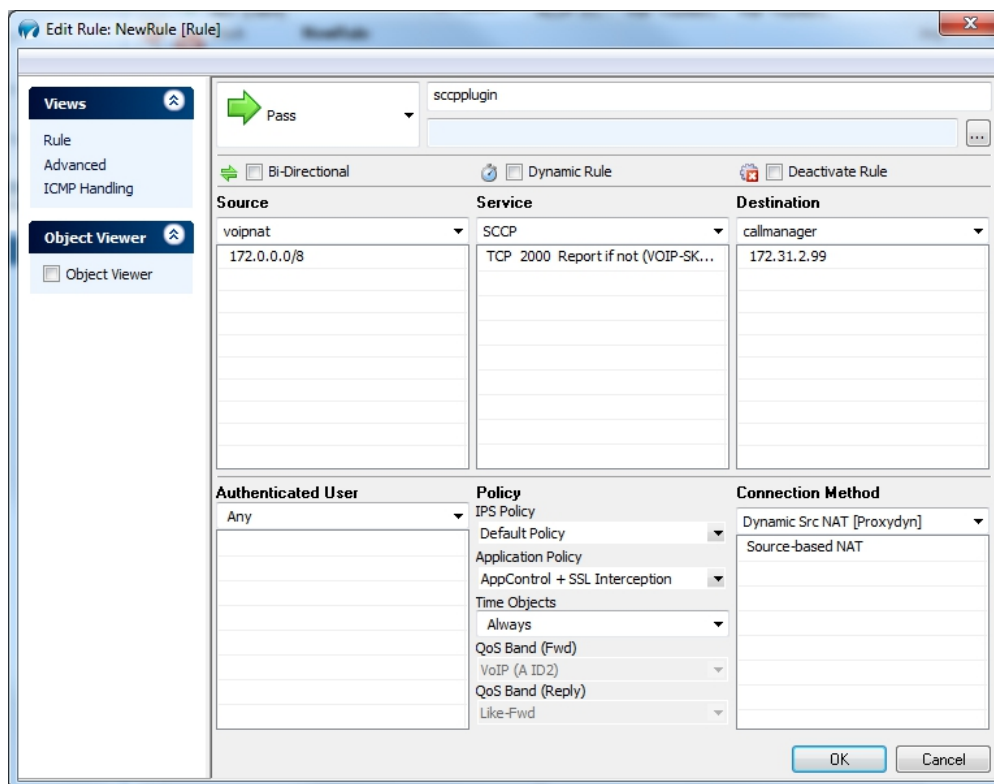
Original	Translated	PArp
172.31.2.0/8	192.168.2.0/8	
172.31.10.0/8	192.168.10.0/8	

In a call setup message the real address of the phone is translated to the virtual address. As soon as the other participant of the call receives the modified call setup message it starts sending its voice

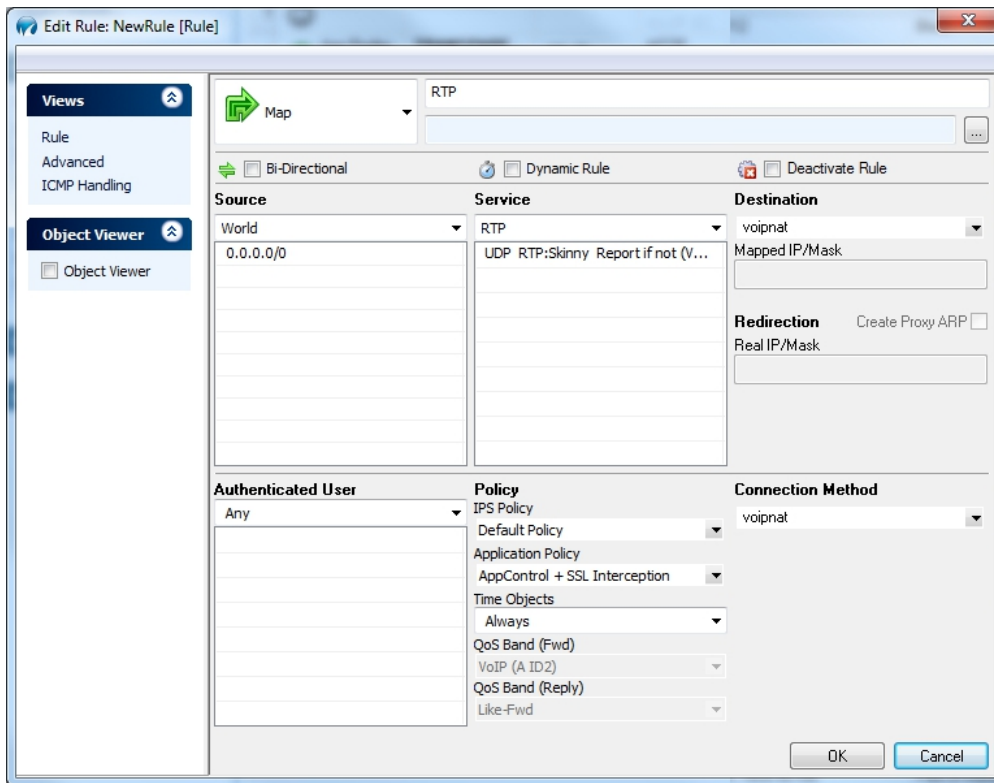
stream to the virtual address of the peer. The firewall next to the receiver of the media stream re-translates the virtual IP address back to the real address of the participant.

The firewall rule required for proper address translation handling has to contain a reference to the service object with the *RTP Dyn*. Service label specified in the skinny plugin (see above). The mapping rule action controls how the address mapping is performed. To use the same address map which is used by the skinny plugin, select the same map in the **Redirection** and **Source Translation** section. If no address translation is required then the [Pass](#) firewall action is to be used.

Skinny signal protocol firewall rule with Skinny firewall plugin:



RTP firewall rule with network address translation from the voipnat address translation map:



Figures

1. skinny_tcp.jpg
2. skinny_srv.jpg
3. hub_ip.jpg
4. transl_fw.jpg
5. sccp_plugin_new.jpg
6. rtp_map_new.jpg

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