

**Hardware and Software**  
Engineered to Work Together



## Cisco UCM 4.x H.323 Interworking to AT&T SIP with Acme Packet 3000-4000 Series SBC

A Technical Application Note



## Disclaimer

The following is intended to outline our general product direction. It is intended for information purposes only, and may not be incorporated into any contract. It is not a commitment to deliver any material, code, or functionality, and should not be relied upon in making purchasing decisions. The development, release, and timing of any features or functionality described for Oracle's products remains at the sole discretion of Oracle.

### Emergency 911/E911 Services Limitations

While AT&T IP Flexible Reach services support E911/911 calling capabilities in certain circumstances, there are significant limitations on how these capabilities are delivered. Please review the AT&T IP Flexible Reach Service Guide in detail to understand these limitations and restrictions.

### Specific IP endpoints Support

Specific IP endpoints are supported with IP Flex Reach. These endpoints must support SCCP and NTE. The Cisco IP endpoints that support SCCP and NTE are:

VG224 - 7902, 7905, 7911, 7912, 7931, 7937, 7940, 7941, 7942, 7945, 7960, 7961, 7962, 7965, 7970, 7971, 7975

### Future new phone models

The Cisco IP endpoints that do NOT support NTE and thus are NOT supported with IP Flex Reach are: 7910, 7920, 7935, 7936

VG248 - DPA-7610, DPA-7630

Please refer to the following Cisco website for further information.

[http://www.cisco.com/en/US/docs/voice\\_ip\\_comm/cucm/srnd/6x/media.html#wp1055031](http://www.cisco.com/en/US/docs/voice_ip_comm/cucm/srnd/6x/media.html#wp1055031)

HIPCS is not supported with Cisco Unified Communication Manager

If the customer is in a HIPCS serving area, Cisco Unified Communication Manager SIP is not supported. Please consult with your customer care or sales person to determine if you are in a HIPCS serving area.

AT&T IP Teleconferencing Service is not supported when G.729 is configured on Cisco Unified Communication Manager

Cisco Unified Communication Manager only supports a single codec on an IP trunk. Since the AT&T IP Teleconferencing (IPTC) Service supports G.711, a Cisco Unified Communication Manager configured for G.729 will not work with the IPTC service.

Abstract.....	2
Introduction .....	2
Intended Audience .....	3
Support .....	3
Design Goals.....	3
IWF Peering Scenario with UCM 4.x .....	4
Notes on Reference Configuration.....	5
Normative References.....	6
Authors' Address .....	6
Full Copyright Statement.....	6
Appendix A: Reference Configuration.....	7

## Abstract

This application note defines an interworking configuration model suitable for the Oracle Communications Session Border Controllers connecting H.323 Trunks from Cisco Unified Communications Manager (UCM) to AT&T's IP Flex Reach service. The reference configuration presented was tested in AT&T's lab.

## Introduction

Within the heterogeneous IP telephony networks deployed today, Enterprises cannot always migrate quickly from legacy H.323 IP-PBX trunks to SIP. Many have found it valuable to use the protocol interworking function (IWF) of the Oracle Communications Session Border Controller for passing VoIP traffic between their network and SIP carriers. The Oracle Communications Session Border Controller supports interworking capabilities between H.323 and SIP protocols. It enables an enterprise to take advantage of carrier SIP trunks while migrating to SIP at a slower pace on their IP-PBX. While transport of media is handled via RTP for both signaling protocols, interworking is required to setup the correct SDP and corresponding H.245 capabilities exchange.

The Cisco UCM (formerly Call Manager) is commonly deployed with version 4.1.x and as such, the protocol of choice for the phones is SCCP (Skinny) and for IP Trunks it has been H.323. This has been the best combination of protocols for access to the widest range of features and the ability to support toll by-pass over the corporate WAN. This also poses a number of problems when an Enterprise is looking to take advantage of service provider SIP trunking cost savings and disaster recovery benefits that can be realized over existing PSTN/PRI technology.

Since there is a protocol mismatch between the UCM H.323 implementation and carrier SIP trunks, a protocol interworking function is needed. There are also special considerations that need to be taken into account because of the manner in which SCCP negotiates the media stream at the end of the call setup sequence as opposed to the beginning. The resulting H.323 signaling uses the "SlowStart" method where the media channels are negotiated after the call Connect message.

Simply interworking H.323 SlowStart to SIP signaling without special consideration of the media negotiation will lead to a SIP INVITE request without SDP for RTP negotiation. The first negotiation of "delayed" SDP Offer in this case is typically in the 200OK call Connect message with the SDP Answer in the ACK to the 200OK. In between the 200OK and the ACK are the SlowStart H.323 and SCCP Connect messages and media setup. This delay can lead to clipping at the beginning of a call, most notably to an IVR or voice mail server. The Cisco solution requires MTP DSP resources in order to present the media offer at the call setup which can become very costly as it requires many servers for large scale deployments.

The SBC has the ability to interwork the UCM H.323 signaling to SIP and include SDP in the INVITE for "Early Offer" without the need for MTP resources.

This Oracle Technical Application Note outlines the recommended configuration for the Acme Packet 3000/4000 series Session Border Controllers, the industry leading Session Border Controllers, for

connecting H.323 Trunks from Cisco UCM 4.x to AT&T's IP Flex Reach service. This document is based on the Acme Packet OS version C5.1.0 but is applicable to images 4.x and higher.

## Intended Audience

This document is intended for use by Oracle Systems Engineers, third party Systems Integrators, and end-users of the Oracle Communications Session Border Controller. It assumes that the reader is familiar with basic operations of the Oracle Communications Session Border Controller.

## Support

Cisco Call Manager Guide is located at the following AT&T URL:

<http://www.corp.att.com/dna/support/>

\*\*The AT&T website is password protected. The ID and Password are provided to each customer when they place an order for IP Flexible Reach or IP Toll Free service.

## Design Goals

The reference configuration represents the most common Cisco UCM (Cisco Call Manager) deployment model: UCM originating H.323 traffic and terminating to AT&T Flex Reach basic SIP trunk via the Oracle SBC. The config also supports bi-directional call-flows (H.323 to SIP and SIP to H.323) via Local-Policy routes. While not presented here, the configuration can be extended to add SIP to SIP Trunking traffic with UCM, which is a common migration strategy for most Enterprise UCM customer networks deploying the Oracle SBC [3].

There is considerable flexibility in configuring H.323 on the SBC. Possible modes include the h323-stack registering as a gateway, acting as a static gateway or behaving as a peering gatekeeper. There are also multiple signaling methods (i.e. FastStart, SlowStart) within the SBC H.323/IWF configuration. This document recommends optimized settings for H.323 configurations whenever possible. Some of the h323-stack and or session-agent parameters MAY need to be modified for interoperability reasons.

This document will annotate each configuration with information on its general applicability. The intent is to:

- Minimize UCM H.323 interoperability issue's by standardizing field configurations
- Provide guidelines for new users for the Session Border Controller
- Provide a configuration template, baselining the H.323 to SIP IWF configuration (with accompanying diagram)
- Flexibility: how resilient the configuration is and how adaptable the configuration is when turning up new UCM H.323 to SIP networks
- Performance: minimize the use of unnecessary configuration objects

## IWF Peering Scenario with UCM 4.x

This section includes a reference architecture diagram, where the Session Border Controller is integrated as an Enterprise CPE trunking Session Border Controller, performing interworking between the Enterprise (UCM) and the AT&T SIP trunk. This reference architecture must be confirmed or modified by the customer according to the specific project requirements.

The Enterprise H.323 UCM peers with the long distance SIP provider via the Oracle Communications Session Border Controller IWF function. Below is the network diagram and representative call-flow.

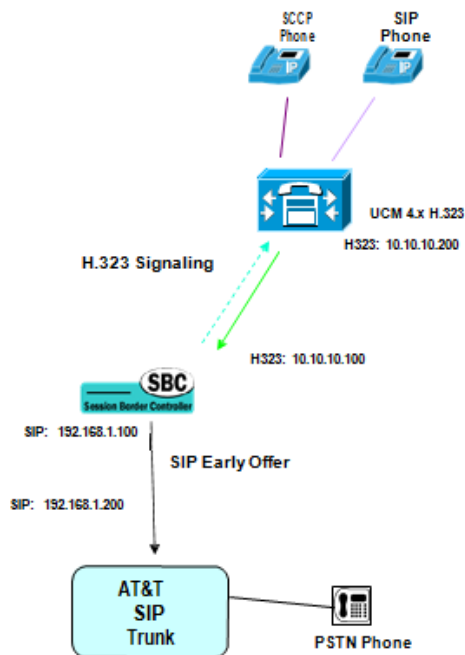


Diagram 1: Enterprise H.323 UCM 4.x to AT&T SIP Trunk via Oracle Communications Session Border Controller IWF

## Cisco CCM H.323 Slow Start GW-GW IWF Call Setup via Oracle Communications Session Border Controller

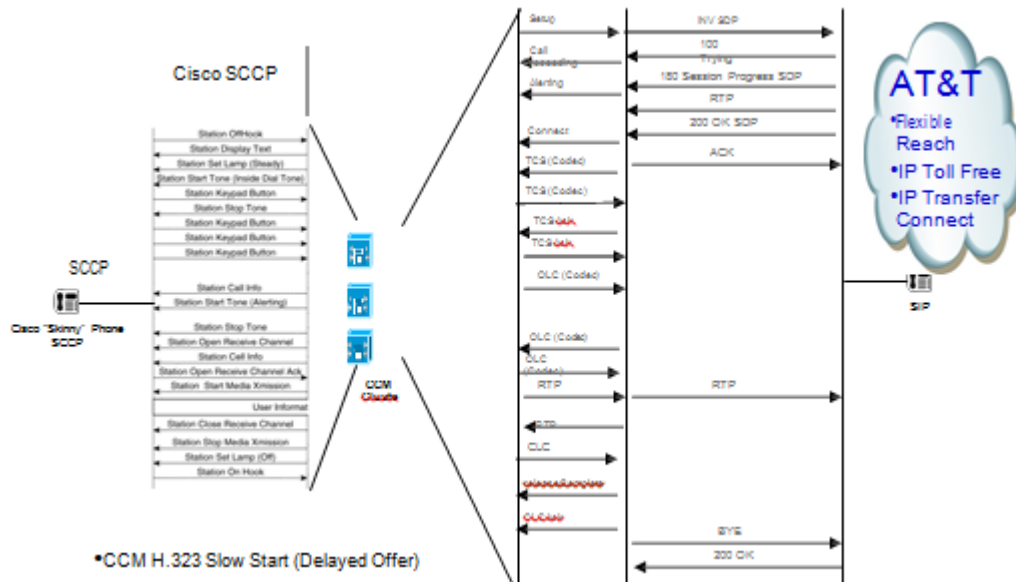


Diagram 2: Call-Flow for Enterprise UCM H.323 4.x to AT&T SIP Trunk via Oracle Communications Session Border Controller IWF

## Notes on Reference Configuration

The Enterprise UCM is configured for H.323 Trunking and acts as a H.323 gateway. The Oracle Communications Session Border Controller is configured to perform IWF function, converting H.323 to SIP signaling between the Enterprise and AT&T's network.

Enabling the SBC's interworking implementation is done via the `iwf-config` element. For H.323 SlowStart calls, it is required to configure the `media-profiles` parameter to provide a list of codecs that will be offered in the SIP SDP exchange. The SBC also requires the customary configuration of H.323 and SIP config objects to support IWF call-flows. H.323 endpoints (gateways, gatekeepers) should be configured as session-agents so that SIP to H.323 calls are correctly interworked.

The realm labeled `peer-h323` and its corresponding `h323-stack` (address 10.10.10.100) is where the H.323 signaling and RTP will enter/exit the SBC. The Enterprise UCM needs to signal to this target IP address.

The realm labeled `core-sip` and its corresponding `sip-interface` (address 192.168.1.100) is where the SIP signaling will enter/exit the SBC to/from the long distance SIP provider.

The Local-Policy configurations route sessions to/from the Enterprise H.323 UCM to the long distance SIP provider.

## WF Peering Configuration

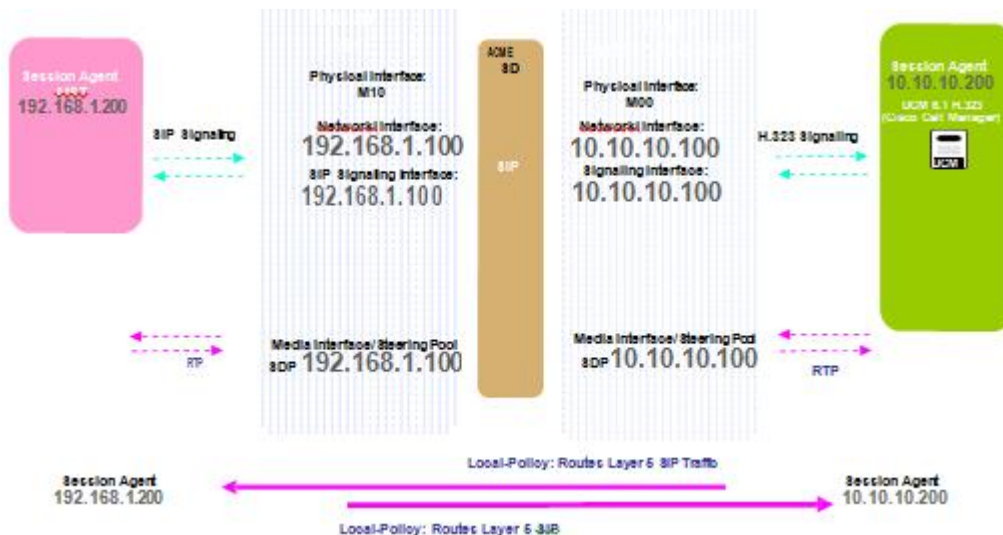


Diagram 3: Network Diagram for Reference Configuration

## Normative References

- [1] Acme Packet, “Net-Net 4000 S-C6.1.0 CLI Configuration Guide”, 400-0061-61, Jan 2009.
- [2] Heyphets, S., “BCP - H.323-SIP Interworking (IWF) Peering Configuration”, 520-0031-00, Aug 2008
- [3] Walker, C., “Cisco UCM 6.x SIP to AT&T SIP with Net-Net 3000-4000 Series SD Application Note”, 510-0014-00, June 2009

## Authors’ Address

Nate Denbow  
 email: [nathan.denbow@oracle.com](mailto:nathan.denbow@oracle.com)

Oracle  
 100 Bedford Drive  
 Bedford, MA 01730



## Appendix A: Reference Configuration

```

acmepacket# show run
h323-config
    state enabled
    log-level WARNING
    response-time 4
    connect-time 180
    cfc2833-payload 101
    alternate-routing proxy
    codec-fallback disabled
    options noReInvite

h323-stack
    name h323-stackccm
    description
    state enabled
    isgateways disabled
    realm-id enterprise-core-ccm
    assoc-stack
    local-ip 10.10.10.100
    max-calls 1000
    max-channels 6
    registration-ssl 120
    terminal-alias h323-ID=acme

    prefixes
    csa-port 1719
    auto-sk-discovery enabled
    multicast 0.0.0.0:0
    gatekeeper 0.0.0.0:0
    sk-identifier
    q931-port 1720
    alternate-transport
    q931-max-calls 200
    h245-tunneling enabled
    fs-in-first-msg disabled
    call-start-fast enabled
    call-start-slow disabled
    media-profiles
    process-registration disabled
    allow-anonymous all
    options inhandTone
        suppress100rel

    proxy-mode alerting
    h245-stage 0
    q931-start-port 0
    q931-number-ports 8192
    dynamic-stack-port 1024
    dynamic-number-ports transparent
    filename
    tcp-keepalive disabled

```

```

ivf-config
  state enabled
  media-profiles
    g729
    PCMU
  logging enabled
  add-reason-bdr enabled
  no-sdp-in-invite disabled

local-policy
  from-address +
  to-address +
  source-realm peer
  activate-time N/A
  deactivate-time N/A
  state enabled
  policy-priority none
  policy-attributes
    next-hop 10.10.10.200
    realm enterprise-core-gom
    action none
    terminate-recursion disabled
    carrier
    start-time 0000
    end-time 2400
    days-of-week U-S
    cost 0
    app-protocol H323
    state enabled
    media-profiles

local-policy
  from-address +
  to-address +
  source-realm enterprise-core-gom
  activate-time N/A
  deactivate-time N/A
  state enabled
  policy-priority none
  policy-attributes
    next-hop SAG:ATT
    realm peer
    action none
    terminate-recursion disabled
    carrier
    start-time 0000
    end-time 2400
    days-of-week U-S
    cost 0
    app-protocol SIP
    state enabled
    media-profiles

media-manager
  state enabled
  latching disabled
  flow-time-limit 86400

```

```

initial-guard-timer          43200
subsq-guard-timer           43200
tcp-flow-time-limit         86400
tcp-initial-guard-timer     300
tcp-subsq-guard-timer       300
tcp-number-of-ports-per-flow 2
hnt-rtcp                    disabled
algd-log-level              NOTICE
mbcd-log-level              NOTICE
options                      min-signal-duration=100
red-flow-port               1985
red-mgcp-port               1986
red-max-trans               10000
red-sync-start-time         5000
red-sync-comp-time         1000
media-policing              enabled
max-signaling-bandwidth     10000000
max-untrusted-signaling     100
min-untrusted-signaling     30
app-signaling-bandwidth     0
tolerance-window           30
rtcp-rate-limit            0
min-media-allocation        32000
min-trusted-allocation      1000
deny-allocation             1000
anonymous-sdp               disabled
arp-msg-bandwidth           32000
fragment-msg-bandwidth      0
rfc2833-timestamp           enabled
default-2833-duration       100
rfc2833-end-pkts-only-for-non-sig disabled
translate-non-rfc2833-event disabled

network-interface
name                          M00
sub-port-id                   0
description                   AT&T/Peer Facing
hostname
ip-address                    192.168.1.100
pri-utility-addr              192.168.1.101
sec-utility-addr              192.168.1.102
netmask                       255.255.255.0
gateway                       192.168.1.1
sec-gateway
gw-heartbeat
state                          disabled
heartbeat                      0
retry-count                    0
retry-timeout                  1
health-score                   0

dns-ip-primary
dns-ip-backup1
dns-ip-backup2
dns-domain
dns-timeout                   11
hip-ip-list                   192.168.1.100
ftp-address

```

```

icmp-address snmp-
address telnet-
address
network-interface
  name M10
  sub-port-id 0
  description Enterprise/Core Facing
  hostname
  ip-address 10.10.10.100
  pri-utility-addr 10.10.10.101
  sec-utility-addr 10.10.10.102
  netmask 255.255.255.0
  gateway 10.10.10.1
  sec-gateway
  gw-heartbeat
    state disabled
    heartbeat 0
    retry-count 0
    retry-timeout 1
    health-score 0
  dns-ip-primary
  dns-ip-backup1
  dns-ip-backup2
  dns-domain
  dns-timeout 11
  hip-ip-list 10.10.10.100
  ftp-address
  icmp-address 10.10.10.100
  snmp-address
  telnet-address

phy-interface
  name M00
  operation-type Media
  port 0
  slot 0
  virtual-mac
  admin-state enabled
  auto-negotiation enabled duplex-mode
  speed

phy-interface
  name M10
  operation-type Media
  port 0
  slot 1
  virtual-mac
  admin-state enabled
  auto-negotiation enabled
  duplex-mode
  speed

realm-config
  identifier peer
  addr-prefix 0.0.0.0

```

```

network-
interfaces M00:0
sm-in-realm enabled
sm-in-network enabled
sm-same-ip enabled
sm-in-system enabled
bw-cac-non-sm disabled
sm-release disabled
qos-enable disabled
max-bandwidth 0
ext-policy-svc
max-latency 0
max-jitter 0
max-packet-loss 0
obsrv-window-size 0
parent-realm
dns-realm
media-policy
in-translationid
out-translationid
in-manipulationid
out-manipulationid
class-profile
average-rate-limit 0
access-control-trust-level
invalid-signal-threshold 0
maximum-signal-threshold 0
untrusted-signal-threshold 0
deny-period 30
symmetric-latching disabled
pat-strip disabled
trunk-context
early-media-allow
additional-prefixes
restricted-latching none
restriction-mask 32
accounting-enable enabled
user-cac-mode none
user-cac-bandwidth 0
user-cac-sessions 0
net-management-control disabled
delay-media-update disabled

realm-config
identifier enterprise-core-ccm
addr-prefix 0.0.0.0
network-interfaces
M10:0
sm-in-realm enabled
sm-in-network enabled
sm-same-ip enabled
sm-in-system enabled
bw-cac-non-sm disabled
sm-release disabled
qos-enable disabled
max-bandwidth 0
ext-policy-svc

```

```

max-latency 0
max-jitter 0
max-packet-loss 0
observed-window-size 0
parent-realm dns-
realm media-policy
in-translationid
out-translationid
in-manipulationid
out-manipulationid
class-profile
average-rate-limit 0
access-control-trust-level
invalid-signal-threshold 0
maximum-signal-threshold 0
untrusted-signal-threshold 0
deny-period 30
symmetric-latching disabled
pai-strip disabled
trunk-context
early-media-allow
additional-prefixes
restricted-latching none
restriction-mask 32
accounting-enable enabled
user-pac-mode none
user-pac-bandwidth 0
user-pac-seconds 0
net-management-control disabled
delay-media-update disabled

```

### sessi agent

hostname	192.168.1.20		
ip-address			
port	5060		
state	enabled		
app-protocol	SIP		
app-type			
transport-method	UDP		
realm-id	peer		
description	AT&T Session Agent Primary		
carriers			
allow-next-hop-ip	enabled		
constraints	enabled		
max-sessions	0		
max-inbound-sessions	0		
max-outbound-sessions	0		
max-burst-rate	0		
max-inbound-burst-rate	0		
max-outbound-burst-rate	0		
max-sustain-rate	0		
max-inbound-sustain-rate	0		
max-outbound-sustain-rate	0		
min-seizures	5		
min-asr	0		
time-to-resume	0		

```

in-service-response 300
in-service-redirect 0
burst-rate-window 0
sustain-rate-window 0
sip-traffic-mode None
proxy-mode
codec-activation
log-logging enabled
send-media-session enabled
response-map
ping-method OPTIONS_homa-0
ping-interval 300
ping-in-service-response-
codes out-service-response-
codes media-profiles
in-transformation-id
out-transformation-id
trust-mode enabled
send-early-media
sip-transport
local-response-map
ping-to-remote-part
ping-to-remote-part
listen-mode disabled
in-manipulation-id
out-manipulation-id
p-asserted-id
trunk-group
max-assistant-contains-rate 0
early-media-allow
invalidate-connections disabled
efc2833-mode none
efc2833-payload 0
codec-policy

session-agent
hostname 192.168.1.201
ip-address
port 5060
state enabled
app-protocol SIP
app-type
transport-method UDP
call-id peer
description AT&T Session Agent Secondary
carriers
allow-next-hop-id enabled
constraints enabled
max-sessions 0
max-inbound-sessions 0
max-outbound-sessions 0
max-burst-rate 0
max-inbound-burst-rate 0
max-outbound-burst-rate 0
max-contains-rate 0
max-inbound-contains-rate 0
max-outbound-contains-rate 0

```

```

min-conn 5
min-conn 0
time-to-live 0
time-to-live 300
in-service-period 0
burst-rate-window 0
sustain-rate-window 0
security-carrier-mode None
proxy-mode
redirection
late-terminating enabled
send-media-session enabled
response-map
ping-method OPTIONS_hops-0
ping-interval 300
ping-in-service-response-codes
out-service-response-codes
media-profiles, in-
translationid
out-translationid
trustme enabled
early-terminating
local-response-map
ping-to-user-part
ping-from-user-part
list-trustme disabled
in-manipulationid
out-manipulationid
p-assesed-id
trunk-group
max-assisted-sustained-rate 0
early-media-allow
invalidate-requests disabled
rfc2833-mode none
rfc2833-payload 0
codec-policy

session-agent
hostname 10.10.10.200
ip-address
port 1720
state enabled
app-protocol H323
app-type H323-CW
transport-method UDP+TCP
call-id *
description Enterprise H323 UCM
carrier
allow-next-hop-ip enabled
constrained disabled
max-sessions 0
max-inbound-sessions 0
max-outbound-sessions 0
max-burst-rate 0
max-inbound-burst-rate 0
max-outbound-burst-rate 0

```



```

max-sustain-rate 0
max-inbound-sustain-rate 0
max-outbound-sustain-rate 0
min-silence 5
min-rsr 0
time-to-recover 0
time-to-recover 0
in-service-period 0
burst-rate-window 0
sustain-rate-window 0
call-music-codec-mode None
on-hook-mode
redirection
local-routing enabled
send-media-session enabled
response-map
ping-method
ping-interval
ping-in-service-response-code
out-of-service-response-code
media-profiles q729 q711Ulaw64k
in-manipulation-id
out-of-manipulation-id
custom
request-headers disabled
stop-requests
local-response-map
ping-to-accept ping-from-user-part
listen-at 0 disabled
in-manipulation-id
out-manipulation-id
presentation-id
trunk-group
max-registers 0
early-media-allow
invalidate-redirections disabled
rfc2833-mode none
rfc2833-load 0
codec-policy

#session-group
group-name ATT
description
state enabled
app-protocol SIP
strategy RoundRobin
dial
192.168.1.200
192.168.1.201

trunk
group
sig-secure disabled
sig-arg 401,407

#sip-conf
state enabled
operation-mode dialog

```

```

dialog-transparency          enabled
home-callsid                peer
egress-callsid              enterprise-cscc-com
outmode                       Public
registrationdomain
registration-host
registration-port            0
registration-service-code    always
initial-time                 500
max-time                      4000
transfer-expire              32
invite-expire                 180
inactive-dynamic-com         32
enforcement-profile
pcc-method
pcc-interval                  10
pcc-state-delay               PccDiat
pcc-load-weight               1
pcc-session-weight            1
pcc-queue-weight              1
pcc-call-id-lifetime          600
pcc-user-lifetime             3600
sip-port                       1988
ad-adapt-time                 10000
ad-async-start-time           5000
ad-async-comp-time            1000
ad-adapt-on-header            disabled
sip-message-len                0
ad-adapt-match                 disabled
extra-method-state             disabled
echo-feature                   disabled
ad-adapt-session-state         0
registration-cache-limit       0
options                         add-pcc-to-tag-no
                                insert-pcc-header max-udp-
                                length=0
                                set-invoice-rate=100ms

```

```

sip-feature
  name                          100=al
  call                           peer
  support-mode-inbound           Strip
  require-mode-inbound           Reject
  proxy-require-mode-inbound     Pass
  support-mode-outbound          Strip
  require-mode-outbound          Reject
  proxy-require-mode-outbound    Pass

```

```

sip-interface
  state                           enabled
  call-id                          peer
  description
  sip-port
    address                        192.168.1.100
    port                           5060
    transport-protocol             UDP
    class-profile

```

```

allow-anonymous
agents-only
carriers
trans-headers 0
invite-headers 0
max-redirect-counts 0
pcc-mode
redirection
contact-mode none
nat-in-headers none
nat-interval 30
max-out-interval 30
registration-caching disabled
min-agg-expire 300
registration-interval 3600
route-to-register disabled
save-on-network disabled
telco-schema disabled
uri-fqdn-domain
trust-mode all
max-nat-interval 3600
nat-int-increment 10
nat-test-increment 30
sig-dynamic-hdr disabled
status-codes 401,407
port-map-start 0
port-map-end 0
in-manipulation-id
out-manipulation-id Privacy
sip-ua-defaults disabled
operator-identifier
anonymous-privacy none
max-incoming-conns 0
per-ua-sip-max-incoming-conns 0
inactive-connect-timeout 0
unterminated-connect-timeout 0
network-id
out-payload-codesc
default-location-string
charging-headers-mode none
charging-function-address-mode none
ccf-address
ccf-addresses
trans-headers-mode none
implicit-service-codesc disabled
fcs3833-payload 101
fcs3833-mode transparent
const-int-name
#AAGPAACTAG
local-response-map
enforcement-profile
efec-call-transfer disabled
enter-authentication-calls
top-keepsalive none
add-on-headers disabled
add-on-profiles sip-

```

manipulation

```

NAME
description
header-rule
    name
    header-name
    action
    comparison-type
    match-value
    msg-type
    msg-value
    methods
    element-rule
        name
        parameter-name
        type
        action
        match-value-type
        comparison-type
        match-value
        msg-value
header-rule
    name
    header-name
    action
    comparison-type
    match-value
    msg-type
    msg-value
    methods
    element-rule
        name
        parameter-name
        type
        action
        match-value-type
        comparison-type
        match-value
        msg-value
header-rule
    name
    header-name
    action
    comparison-type
    match-value
    msg-type
    msg-value
    methods
    element-rule
        name
        parameter-name
        type
        action
        match-value-type
        comparison-type
        match-value
        msg-value
header-rule
    name
    header-name
    action
    comparison-type
    match-value
    msg-type
    msg-value
    methods
    element-rule
        name
        parameter-name
        type
        action
        match-value-type
        comparison-type
        match-value
        msg-value

```

Privacy  
changing ip

PRI\_Header  
P-Asserted-Identity  
manipulate  
case-sensitive

any

PRI\_Local\_IP  
pri-host  
replace  
any  
case-sensitive

\$LOCAL\_IP

PRI\_Header  
P-Preferred-Identity  
manipulate  
case-sensitive

any

PRI\_Local\_IP  
pri-host  
replace  
any  
case-sensitive

\$LOCAL\_IP

From\_Header  
From  
manipulate  
case-sensitive

request

From\_Header  
pri-host  
replace  
any  
case-sensitive

\$LOCAL\_IP

```

name
header-name
action
comparison-type
match-value
match-type
match-value
method
element-rule
    name
    parameter-name
    type
    action
    match-value-type
    comparison-type
    match-value
    match-value
header-rule
    name
    header-name
    action
    comparison-type
    match-value
    match-type
    match-value
    method
    element-rule
        name
        parameter-name
        type
        action
        match-value-type
        comparison-type
        match-value
        match-value
header-rule
    name
    header-name
    action
    comparison-type
    match-value
    match-type
    match-value
    method
    element-rule
        name
        parameter-name
        type
        action
        match-value-type
        comparison-type
        match-value
        match-value
header-rule
    name
    header-name
    action

```

```

To_Header
To
manipulate
case-sensitive
request
To_header
uri-host
replace
any
case-sensitive
$REMOTE_IP

```

```

RPI_Header
Remote-Party-ID
manipulate
case-sensitive
any
RPI_header
uri-host
replace
any
case-sensitive
$LOCAL_IP

```

```

Referred-Header
Referred-By
manipulate
case-sensitive
any
referredbyhdr
uri-host
replace
any
case-sensitive
$LOCAL_IP

```

```

ReferredTo
Refer-To
manipulate

```

```

comparison-type      case-sensitive
match-value
msg-type            any
match-value
method
element-rule
    name              case-sensitive
    parameter-name
    type              uri-host
    action            replace
    match-real-type   any
    comparison-type   case-sensitive
    match-value
    match-value      $REMOTE_IP

header-rule
    name              ContactHdr
    header-name       Contact
    action            manipulate
    comparison-type   case-sensitive
    match-value
    msg-type          any
    match-value
    method
    element-rule
        name          ContactHostReplace
        parameter-name
        type          uri-host
        action        replace
        match-real-type   any
        comparison-type   case-sensitive
        match-value
        match-value    $LOCAL_IP

steering-pool
    ip-address        192.168.1.100
    start-port        16384
    end-port          32767
    call-id
    network-interface

steering-pool
    ip-address        10.10.10.100
    start-port        16384
    end-port          32767
    call-id           enterprise-cs3-001
    network-interface

system-config
    hostname          Enterprise
    description       Enterprise-SD
    location
    nls-system-contact
    nls-system-name
    nls-system-location
    cscc-enabled      enabled
    enable-cscc-auth-trans   disabled

```

```
enable-smp-sylog-notify      enabled
enable-smp-monitor-traps    enabled
enable-smp-monitor-traps    disabled
smp-sylog-his-table-length   1
smp-sylog-level              WARNING
system-log-level             WARNING
process-log-level            NOTICE
process-log-ip-address      0.0.0.0
process-log-port              0
calltrace                    disabled
internaltrace                 disabled
log-filter                    all
default-gateway              192.168.1.1
contact                       enabled
exceptions
timer-timerout                0
console-timerout              0
remote-control                 enabled
link-adv-adv-adv-adv-adv-adv disabled
```



Cisco UCM 4.x H.323 Interworking to AT&T SIP  
with Acme Packet 3000-4000 Series SBC  
February 2014

Oracle Corporation  
World Headquarters  
500 Oracle Parkway  
Redwood Shores, CA 94065  
U.S.A.

Worldwide Inquiries:  
Phone: +1.650.506.7000  
Fax: +1.650.506.7200

oracle.com



Oracle is committed to developing practices and products that help protect the environment

Copyright © 2014, Oracle and/or its affiliates. All rights reserved.

This document is provided for information purposes only, and the contents hereof are subject to change without notice. This document is not warranted to be error-free, nor subject to any other warranties or conditions, whether expressed orally or implied in law, including implied warranties and conditions of merchantability or fitness for a particular purpose. We specifically disclaim any liability with respect to this document, and no contractual obligations are formed either directly or indirectly by this document. This document may not be reproduced or transmitted in any form or by any means, electronic or mechanical, for any purpose, without our prior written permission.

Oracle and Java are registered trademarks of Oracle and/or its affiliates. Other names may be trademarks of their respective owners.

Intel and Intel Xeon are trademarks or registered trademarks of Intel Corporation. All SPARC trademarks are used under license and are trademarks or registered trademarks of SPARC International, Inc. AMD, Opteron, the AMD logo, and the AMD Opteron logo are trademarks or registered trademarks of Advanced Micro Devices. UNIX is a registered trademark of The Open Group. 0114

**Hardware and Software, Engineered to Work Together**