

Интегрированные функции безопасности и унифицированных коммуникаций. Лучшие практики и рекомендации по настройке. Часть 2

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CISCO *Live!*



Agenda

- WAN / VPN QoS Design
- Integrated Voice. CUBE

The background is a dark blue field filled with numerous small, semi-transparent squares and dots. These elements are scattered across the frame, with a higher concentration of larger squares in the upper left and a trail of smaller dots and squares extending from the upper right towards the bottom right. The colors of these elements include various shades of blue, cyan, green, yellow, orange, and red, creating a vibrant, pixelated effect.

Introduction to Strategic QoS Design




Translating Business-Relevance to QoS Treatments

Apply RFC 4594-based Marking / Queuing / Dropping Treatments



Translating Business-Relevance to QoS Treatments

Apply RFC 4594-based Marking / Queuing / Dropping Treatments

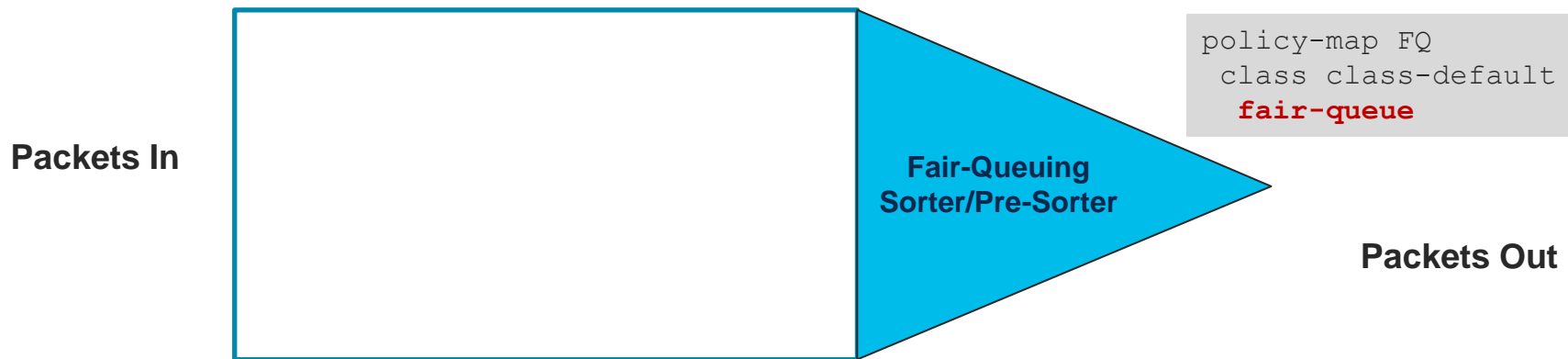
	Application Class	Per-Hop Behavior	Queuing and Dropping	Application Examples
Relevant 	VoIP Telephony	EF	Priority Queue (PQ)	Cisco IP Phones (G.711, G.729)
	Broadcast Video	CS5	(Optional) PQ	Cisco IP Video Surveillance / Cisco Enterprise TV
	Real-Time Interactive	CS4	(Optional) PQ	Cisco TelePresence
	Multimedia Conferencing	AF4	BW Queue + DSCP WRED	Cisco Jabber, Cisco WebEx
	Multimedia Streaming	AF3	BW Queue + DSCP WRED	Cisco Digital Media System (VoDs)
	Network Control	CS6	BW Queue	EIGRP, OSPF, BGP, HSRP, IKE
	Signalling	CS3	BW Queue	SCCP, SIP, H.323
	Ops / Admin / Mgmt (OAM)	CS2	BW Queue	SNMP, SSH, Syslog
	Transactional Data	AF2	BW Queue + DSCP WRED	ERP Apps, CRM Apps, Database Apps
	Bulk Data	AF1	BW Queue + DSCP WRED	E-mail, FTP, Backup Apps, Content Distribution
Default 	Default Forwarding	DF	Default Queue + RED	Default Class
Irrelevant 	Scavenger	CS1	Min BW Queue (Deferential)	YouTube, Netflix, iTunes, BitTorrent, Xbox Live

The background is a dark blue field filled with numerous small, semi-transparent squares and dots in shades of blue and yellow. These elements are scattered across the frame, with a higher concentration of yellow squares and dots on the right side, creating a sense of depth and movement.

WAN Edge QoS Design

QoS Tools Review: Queuing and Dropping Tools

(Flow-Based) Fair-Queuing



A flow is defined by **five matching tuples**:

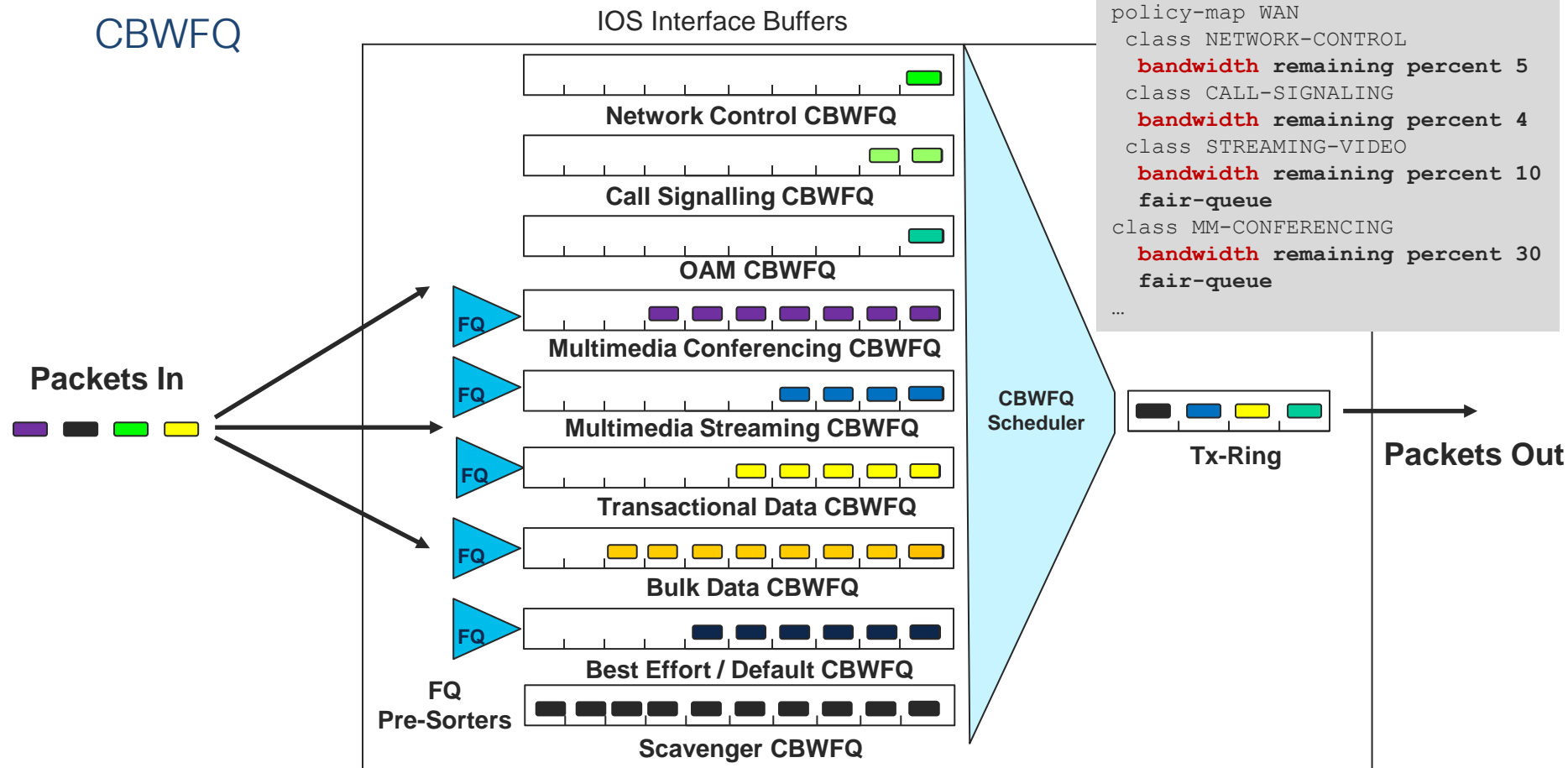
Source Address + Source Port

Destination Address + Destination Port

Layer 4 Protocol (TCP or UDP)

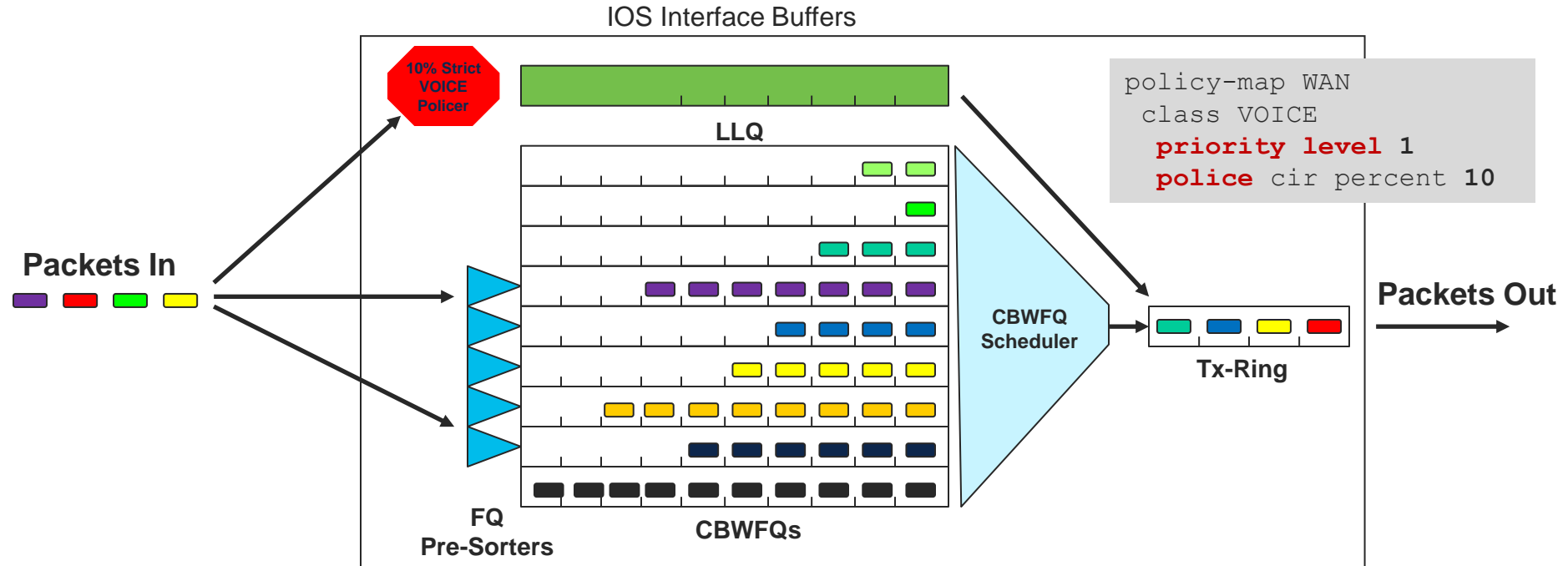
QoS Tools Review: Queuing and Dropping Tools

CBWFQ



QoS Tools Review: Queuing and Dropping Tools

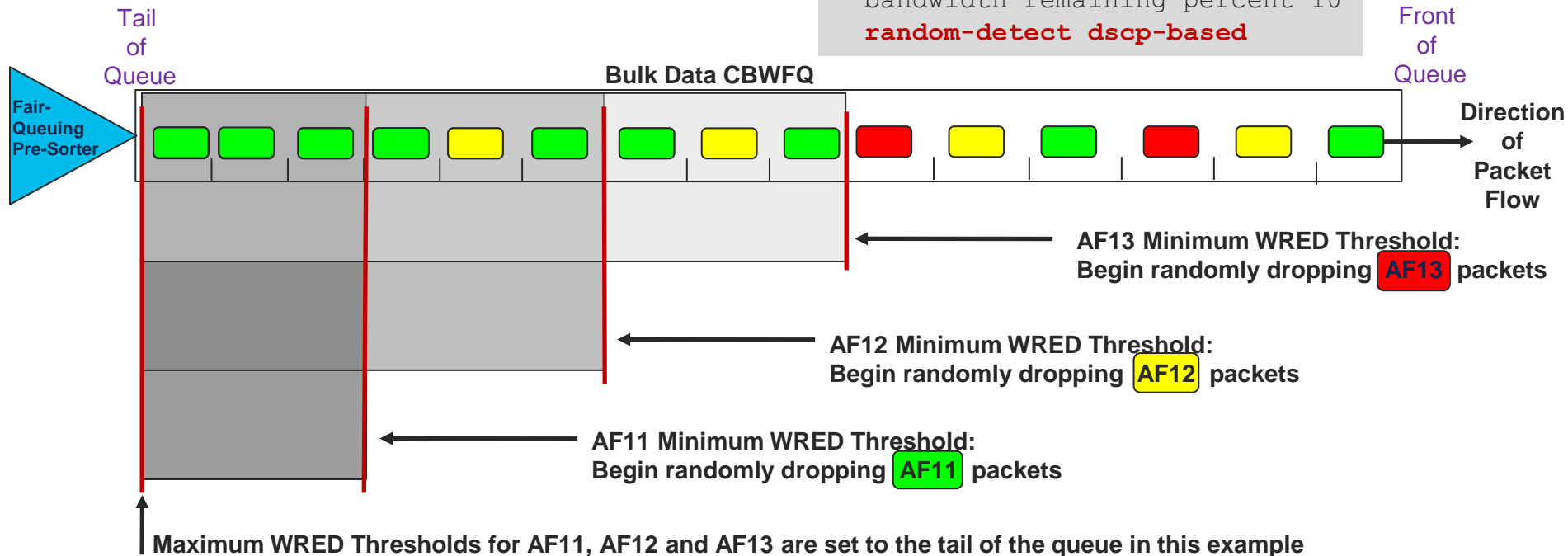
LLQ: Single-LLQ Operation and Configuration



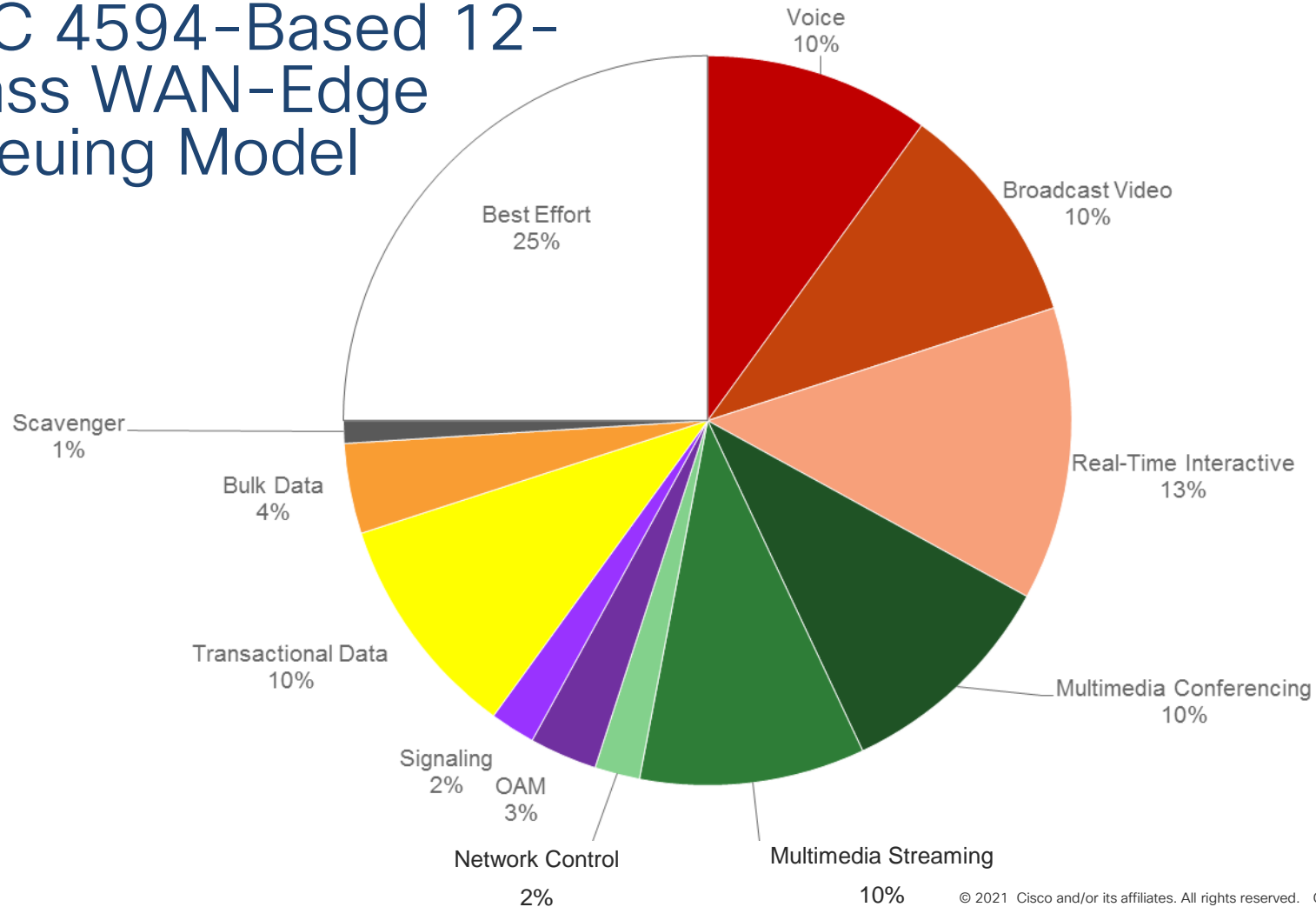
QoS Tools Review: Queuing and Dropping Tools

DSCP-Based WRED

```
policy-map BULK-WRED
class BULK
bandwidth remaining percent 10
random-detect dscp-based
```




RFC 4594-Based 12-Class WAN-Edge Queuing Model



RFC 4594-Based 12-Class Queuing Model Configuration

```
class-map match-all VOICE-DSCP
  match dscp ef
class-map match-all BROADCAST_VIDEO-DSCP
  match dscp cs5
class-map match-all REALTIME_INTERACTIVE-DSCP
  match dscp cs4
class-map match-all NETWORK-CONTROL-DSCP
  match cs6
class-map match-all SIGNALING-DSCP
  match cs3
class-map match-all OAM-DSCP
  match cs2
class-map match-all MULTIMEDIA_CONFERENCING-DSCP
  match dscp af41
class-map match-all MULTIMEDIA_STREAMING-DSCP
  match dscp af31
class-map match-all TRANSACTIONAL-DSCP
  match dscp af21
class-map match-all BULK-DATA-DSCP
  match dscp af11
class-map match-all SCAVENGER-DSCP
  match dscp cs1
```



Note: Appending “-DSCP” to the class-map names distinguishes WAN-Edge egress-queuing class-maps (matching on DSCP values) from the LAN-Edge ingress class-maps (matching via NBAR2).

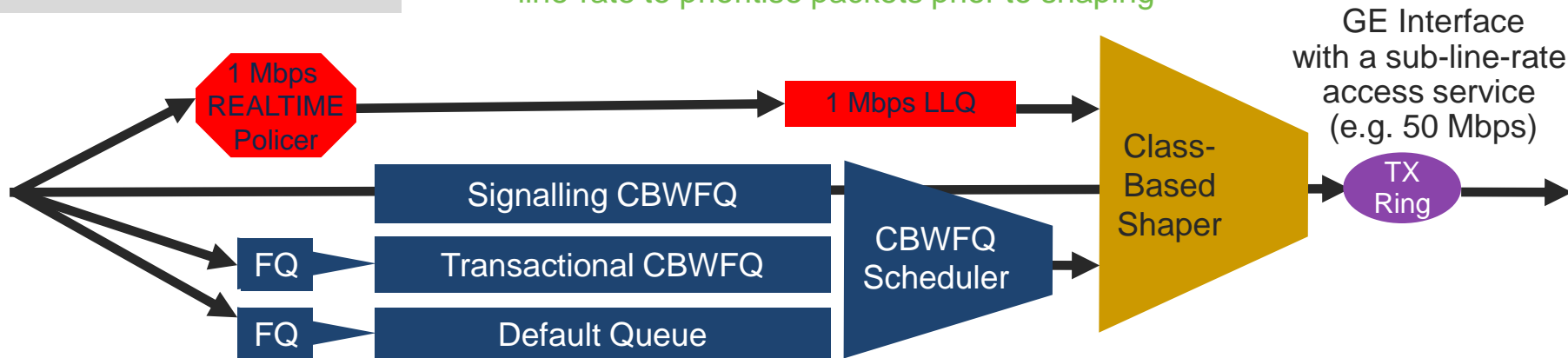
```
policy-map WAN_EDGE-QUEUING
  class VOICE-DSCP
    priority percent 10
  class BROADCAST_VIDEO-DSCP
    priority percent 10
  class REALTIME_INTERACTIVE-DSCP
    priority percent 13
  class NETWORK-CONTROL-DSCP
    bandwidth percent 2
  class SIGNALING-DSCP
    bandwidth percent 2
  class OAM-DSCP
    bandwidth percent 3
  class MULTIMEDIA_CONFERENCING-DSCP
    bandwidth percent 10
    fair-queue
    random-detect dscp-based
  class MULTIMEDIA_STREAMING-DSCP
    bandwidth percent 10
    fair-queue
    random-detect dscp-based
  class TRANSACTIONAL-DSCP
    bandwidth percent 10
    fair-queue
    random-detect dscp-based
  class BULK-DATA-DSCP
    bandwidth percent 4
    fair-queue
    random-detect dscp-based
  class SCAVENGER-DSCP
    bandwidth percent 1
  class class-default
    bandwidth percent 25
    fair-queue
    random-detect dscp-based
```

What Changes for Sub-Line-Rate Interfaces?

```
policy-map QUEUING
class REALTIME
priority 1000
class SIGNALING
bandwidth x
class TRANSACTIONAL
bandwidth y...
class class-default
fair-queue
```

```
policy-map HQoS-50MBPS
class class-default
shape average 50000000
service-policy QUEUING
```

- Queuing policies will not engage unless the interface is congested
- A shaper will guarantee that traffic will not exceed the contracted rate
- A nested queuing policy will force queuing to engage at the contracted sub-line-rate to prioritise packets prior to shaping

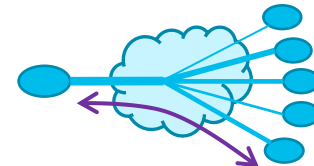




VPN QoS Design

Aggregate Priority Load

IPSec VPN Design Recommendation

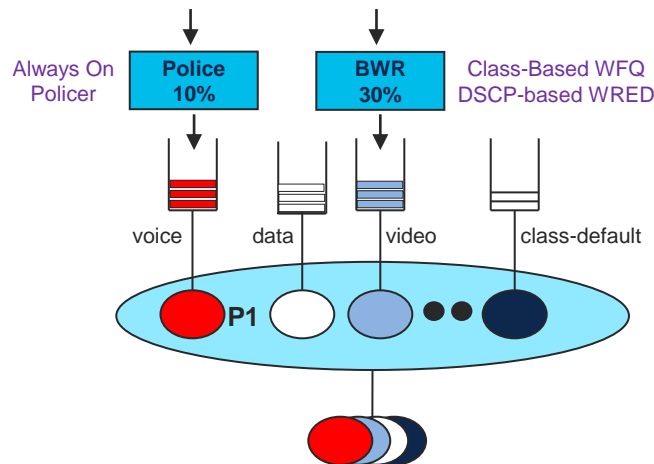


- For Voice, use an **Always On policer**, rather than a Conditional policer

```
class VOICE
  priority level 1
  police cir percent 10
```

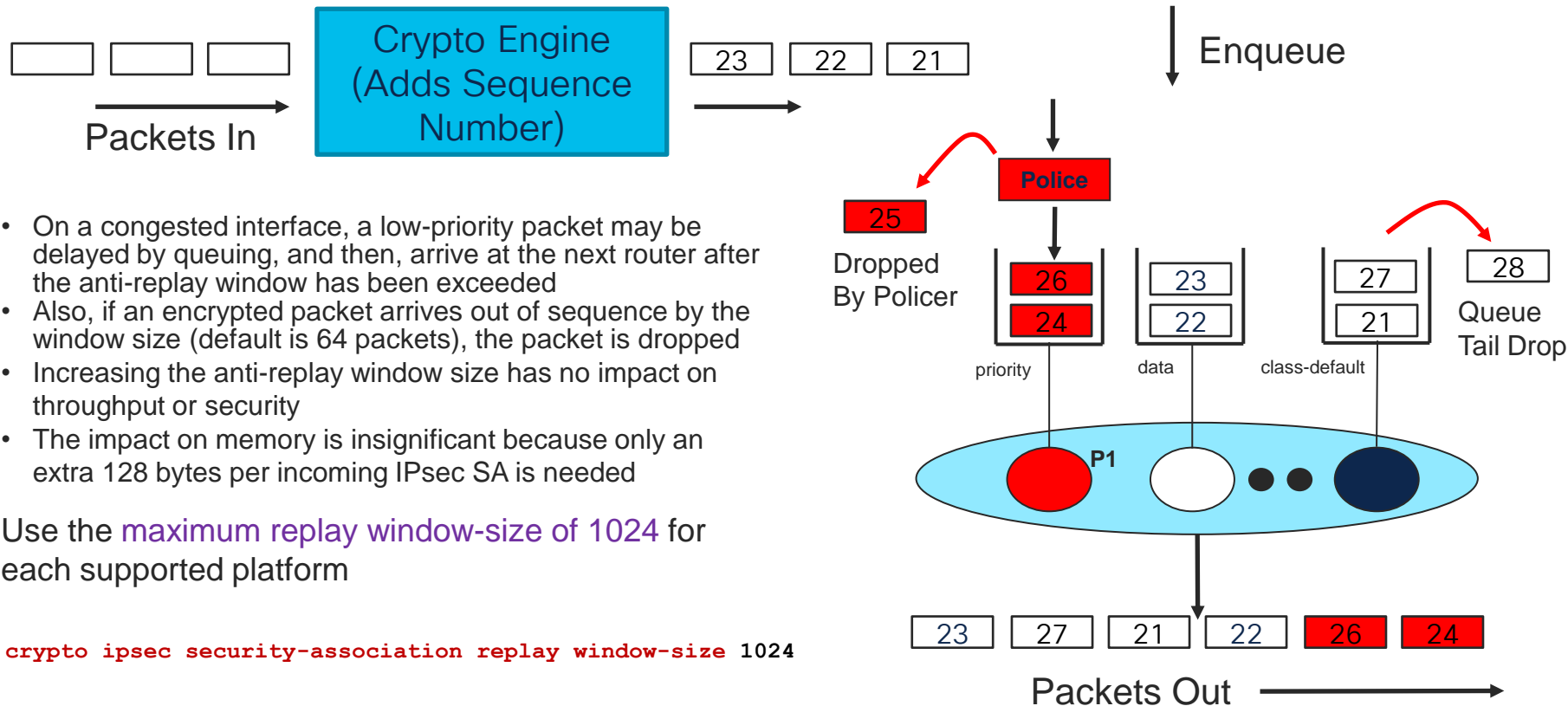
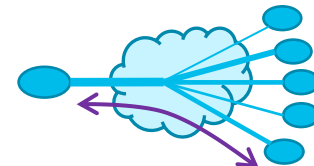
- For Video, use a **Bandwidth Remaining Percent (BWR)** queue with **DSCP-based WRED**, rather than a level 2 Priority queue

```
class INTERACTIVE-VIDEO
  bandwidth remaining percent 30
  random-detect dscp-based
```



IPsec Anti-Replay and QoS

IPSec VPN Design Recommendation



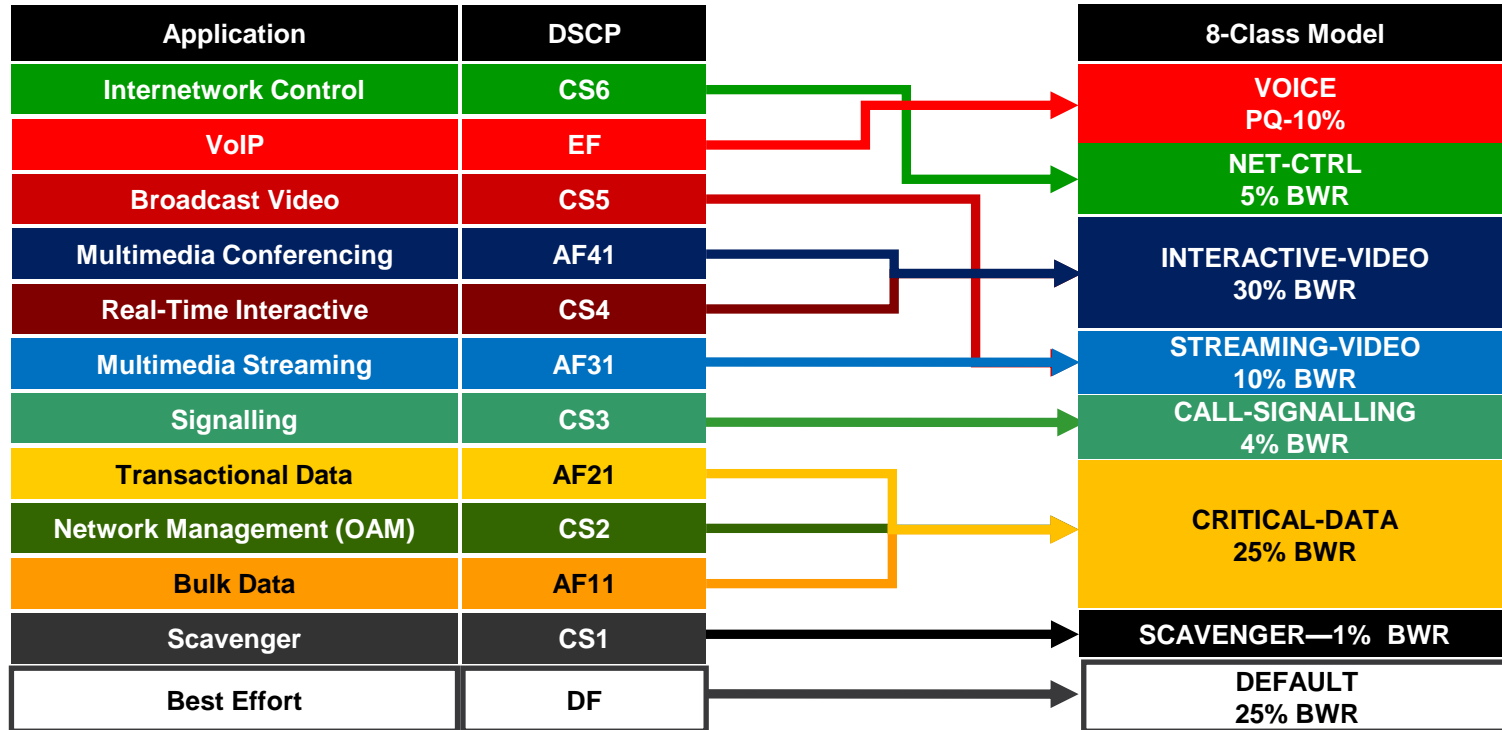
- On a congested interface, a low-priority packet may be delayed by queuing, and then, arrive at the next router after the anti-replay window has been exceeded
- Also, if an encrypted packet arrives out of sequence by the window size (default is 64 packets), the packet is dropped
- Increasing the anti-replay window size has no impact on throughput or security
- The impact on memory is insignificant because only an extra 128 bytes per incoming IPsec SA is needed

Use the **maximum replay window-size of 1024** for each supported platform

`crypto ipsec security-association replay window-size 1024`

IPSec VPN Egress QoS Models

Example: Combining 12 Classes into an 8-Class Model



PQ = Priority Queue
BWR = Bandwidth Remaining

Note: Bandwidth Remaining
Percentages must equal 100%

IPSec VPN 8-Class Egress Queuing Model

Child Queuing Policy

8-Class Queuing Model Class-Maps

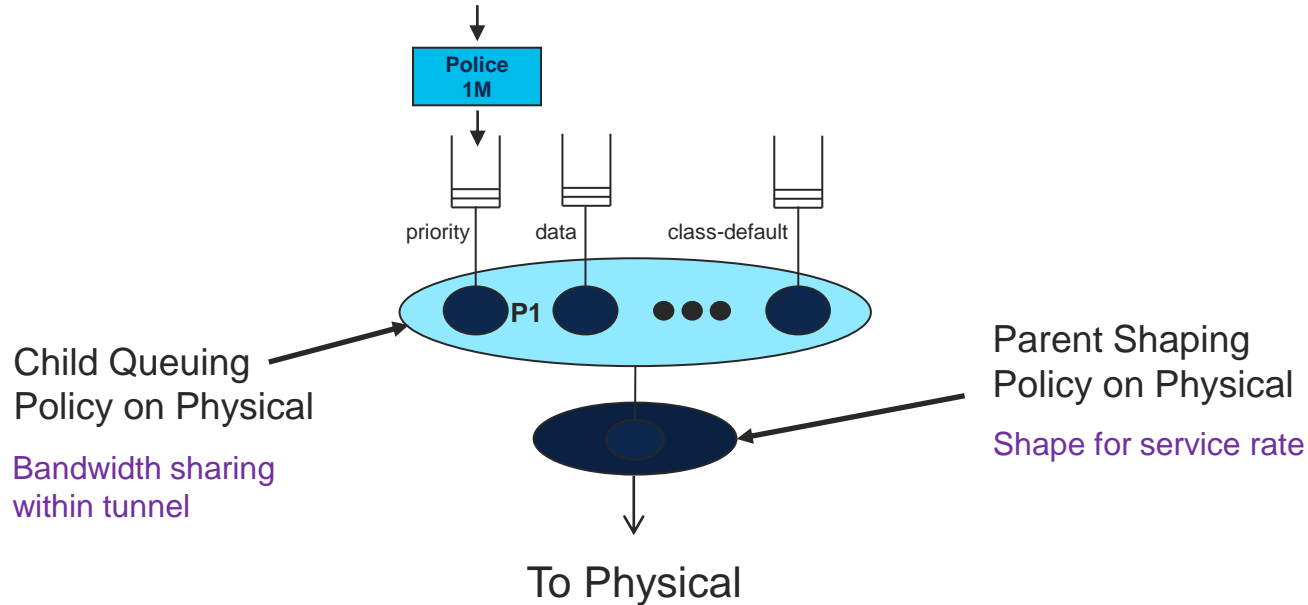
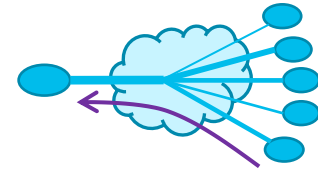
```
class-map match-any VOICE-DSCP
  match dscp ef
class-map match-any INTERACTIVE-VIDEO-DSCP
  match dscp cs4 af41 af42 af43
class-map match-any STREAMING-VIDEO-DSCP
  match dscp cs5 af31 af32 af33
class-map match-any NETWORK-CONTROL-DSCP
  match dscp cs6
class-map match-any SIGNALING-DSCP
  match dscp cs3
class-map match-any CRITICAL-DATA-DSCP
  match dscp cs2 af11 af12 af13 af21 af22 af23
class-map match-any SCAVENGER-DSCP
  match dscp cs1
```

8-Class Queuing Policy-Map

```
policy-map EDGE-QUEUEING
  class VOICE-DSCP
    priority level 1
    police cir percent 10
  class INTERACTIVE-VIDEO-DSCP
    bandwidth remaining percent 30
    random-detect dscp-based
  class STREAMING-VIDEO-DSCP
    bandwidth remaining percent 10
    random-detect dscp-based
  class NETWORK-CONTROL-DSCP
    bandwidth remaining percent 5
  class SIGNALING-DSCP
    bandwidth remaining percent 4
  class CRITICAL-DATA-DSCP
    bandwidth remaining percent 25
    random-detect dscp-based
  class SCAVENGER-DSCP
    bandwidth remaining percent 1
  class class-default
    bandwidth remaining percent 25
    random-detect
```

Branch QoS Scheduling Hierarchy

Two Levels: Child / Parent



Branch QoS Scheduling Hierarchy

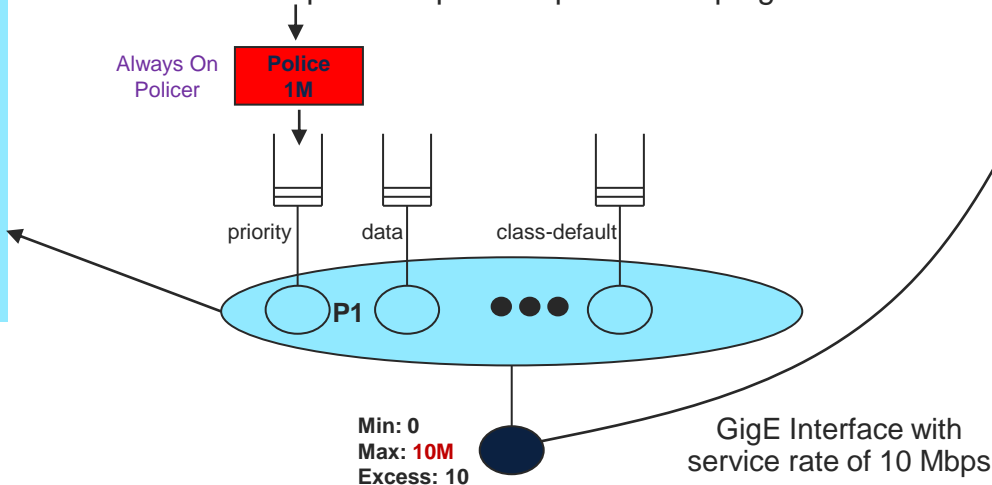
Two Levels: Child / Parent

```
policy-map EDGE-QUEUEING
class INTERACTIVE-VIDEO
  bandwidth remaining percent 30
  random-detect dscp-based
class STREAMING-VIDEO
  bandwidth remaining percent 10
  random-detect dscp-based
class CALL-SIGNALING
  bandwidth remaining percent 4
class NET-CTRL
  bandwidth remaining percent 5
class CRITICAL-DATA
  bandwidth remaining percent 25
  random-detect dscp-based
class SCAVENGER
  bandwidth remaining percent 1
class VOICE
  priority level 1
  police cir percent 10
class class-default
  bandwidth remaining percent 25
  random-detect
```

```
policy-map POLICY-TRANSPORT-1
class class-default
  shape average 10 Mbps
  service-policy EDGE-QUEUEING
```

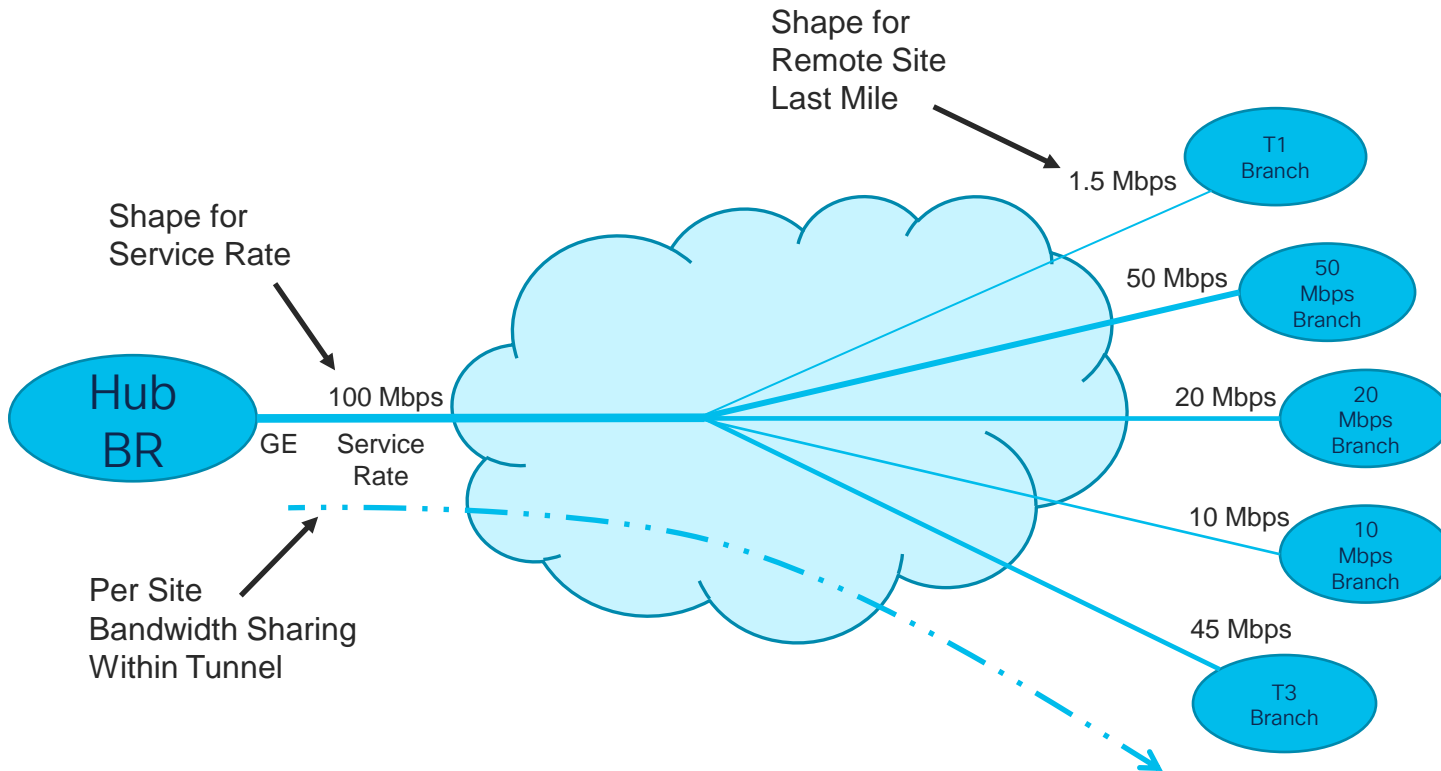
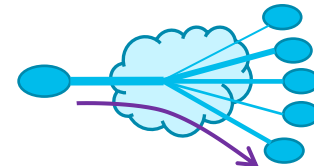
```
interface GigabitEthernet0/0
  bandwidth 10000
  service-policy output POLICY-TRANSPORT-1
```

- A **shaper** will guarantee that traffic will **not exceed the contracted rate**
- A **nested queuing policy** will force queuing to engage at the **contracted sub-line-rate** to prioritise packets prior to shaping



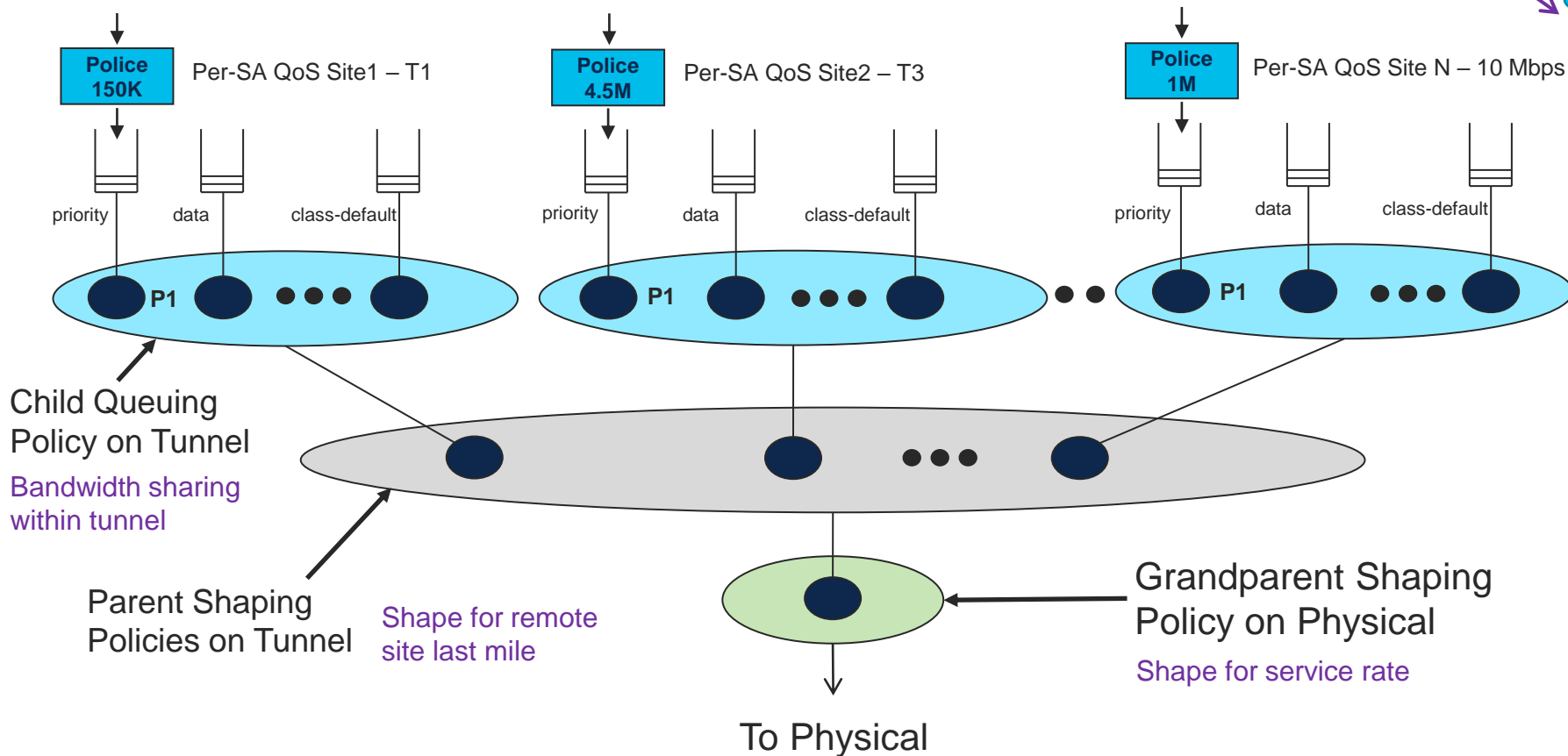
Hub Site QoS Scheduling

Three Levels: Child / Parent / Grandparent



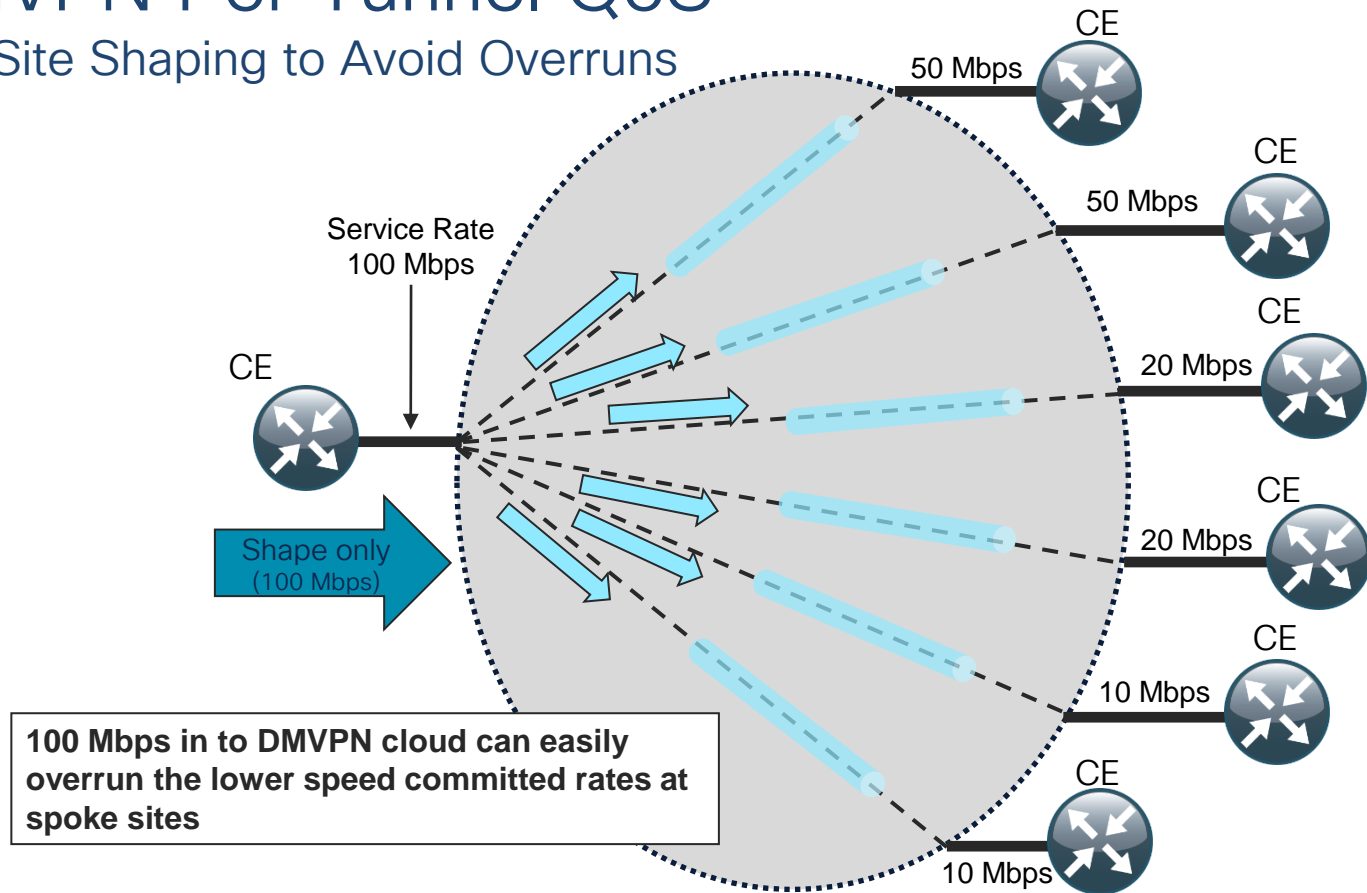
Hub Site QoS Scheduling Hierarchy

Three Levels: Child / Parent / Grandparent



DMVPN Per Tunnel QoS

Per-Site Shaping to Avoid Overruns



DMVPN Hub Per Tunnel QoS

Implementing Per-Site Traffic Shaping

```
policy-map GROUP-50MBPS-POLICY
class class-default
shape average 50 Mbps
bandwidth remaining ratio 50
service-policy EDGE-QUEUEING
```

```
policy-map GROUP-20MBPS-POLICY
class class-default
shape average 20 Mbps
bandwidth remaining ratio 20
service-policy EDGE-QUEUEING
```

```
policy-map GROUP-10MBPS-POLICY
class class-default
shape average 10 Mbps
bandwidth remaining ratio 10
service-policy EDGE-QUEUEING
```

Separate parent shaper policies for each remote-site bandwidth

```
policy-map TRANSPORT-1-SHAPE-ONLY
class class-default
shape average 100 Mbps
!
interface GigabitEthernet0/0/3
bandwidth 100000
service-policy output TRANSPORT-1-SHAPE-ONLY

interface Tunnel10
bandwidth 100000
nhp map group GROUP-10MBPS service-policy output GROUP-10MBPS-POLICY
nhp map group GROUP-20MBPS service-policy output GROUP-20MBPS-POLICY
nhp map group GROUP-50MBPS service-policy output GROUP-50MBPS-POLICY
```

Signal from the spoke to the hub to use the correct policy for each remote site

10 Mbps spoke

20 Mbps spoke

50 Mbps spoke

Remote Site Tunnel Configurations

```
interface GigabitEthernet0/0
bandwidth 100000
service-policy output POLICY-TRANSPORT-1
!
interface Tunnel10
bandwidth 10000
nhp group GROUP-10MBPS
tunnel source GigabitEthernet0/0
tunnel vrf TRANSPORT-1
```

```
interface GigabitEthernet0/0
bandwidth 20000
service-policy output POLICY-TRANSPORT-1
!
interface Tunnel10
bandwidth 20000
nhp group GROUP-20MBPS
tunnel source GigabitEthernet0/0
tunnel vrf TRANSPORT-1
```

```
interface GigabitEthernet0/0
bandwidth 50000
service-policy output POLICY-TRANSPORT-1
!
interface Tunnel10
bandwidth 50000
nhp group GROUP-50MBPS
tunnel source GigabitEthernet0/0
tunnel vrf TRANSPORT-1
```

Per-Tunnel shapers

50 Mbps

50 Mbps

20 Mbps

20 Mbps

10 Mbps

10 Mbps

BRR=50

BRR=50

BRR=20

BRR=20

BRR=10

BRR=10

Service rate shaper

Shape
(100 Mbps)

List all available policies as **map groups** on hub tunnel interface
Add a **class-default shape-only** policy on the hub physical interface for the service rate

Bandwidth Remaining Ratio

Details

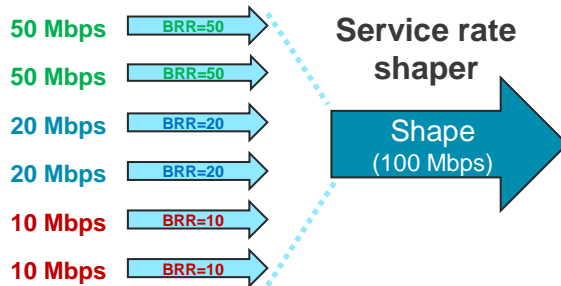


Bandwidth Remaining Ratio (BRR) provides proportional sharing to parent shapers during times of congestion.

If you over-subscribe your hub BR outbound bandwidth with per-tunnel policies that exceed the service rate, the BRR commands on each parent policy means they will get their “fair share” of the remaining bandwidth as compared to the other branch sites.

- If all the per-tunnel BW amounts are 5 Mbps or greater, we use a BRR value of $BW / 1 \text{ Mbps}$. (i.e. 10 Mbps is BRR of 10, 50 Mbps is BRR of 50, etc.)
- If any of the per-tunnel BW values are less than 5 Mbps, we use a BRR value of $BW / 100 \text{ Kbps}$. (i.e. 3 Mbps is BRR of 30, 1.5 Mbps is BRR of 15, etc.)

Per-Tunnel shapers



When the total bandwidth exceeds 100 Mbps, each of the per-tunnel shapers will get their fair share based on their BRR values.

Example:

50 Mbps site gets $50 / 160$ or 31.25%

20 Mbps site gets $20 / 160$ or 12.5%

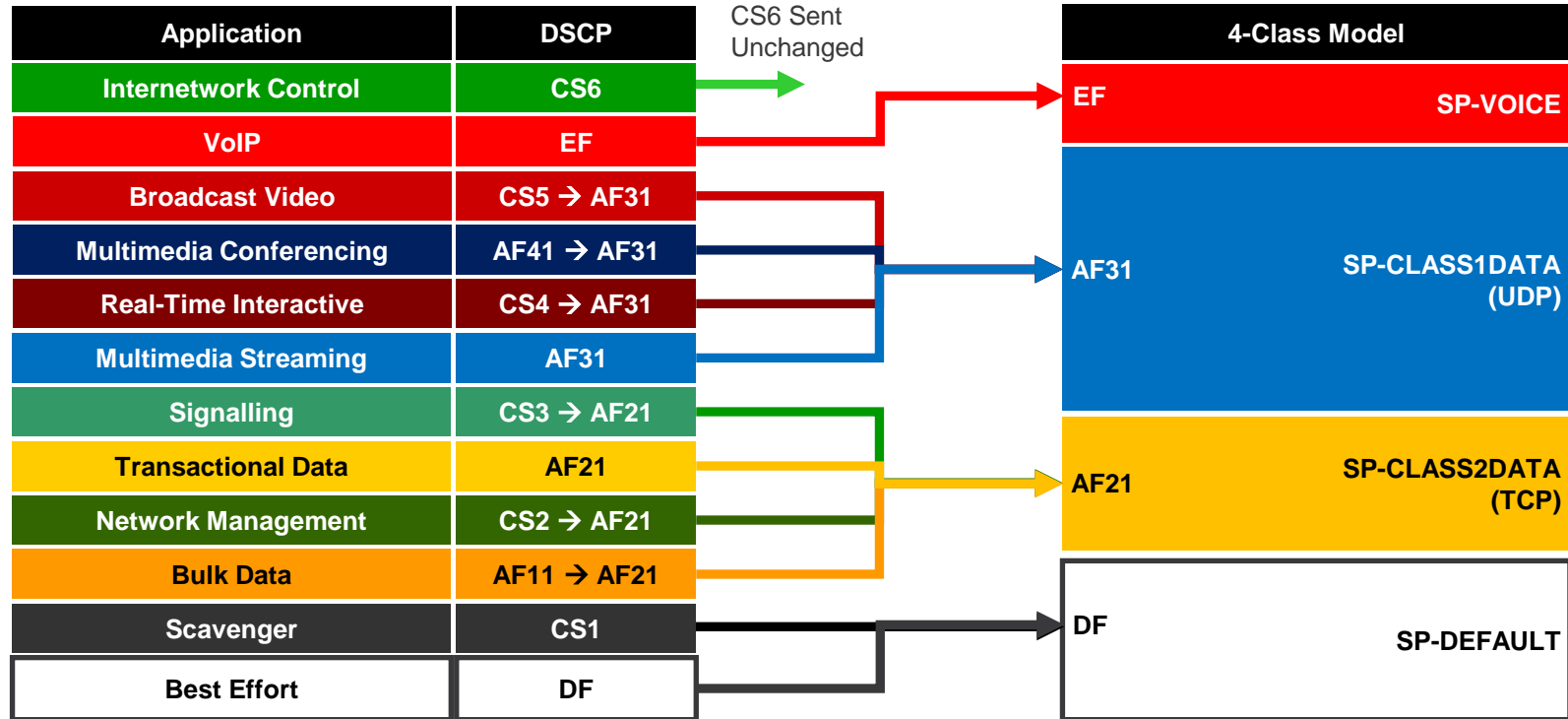
10 Mbps site gets $10 / 160$ or 6.25%

The background of the slide is a dark blue field filled with numerous small, semi-transparent squares and dots in shades of light blue and yellow. These elements are scattered across the entire frame, with a higher density of yellow squares and dots on the right side, creating a sense of depth and movement.

Enterprise to Service- Provider QoS Mapping

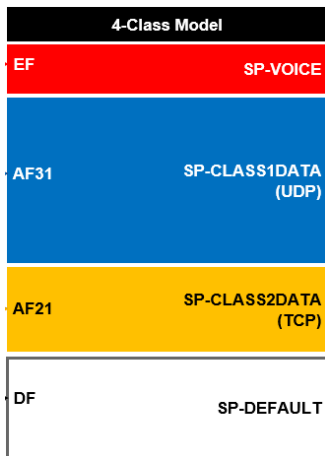
Enterprise to SP Mapping

Example: 4-Class SP Model



4-Class SP QoS Model Configuration

Non-Tunneled Traffic



```
policy-map EDGE-QUEUEING
  class INTERACTIVE-VIDEO
    bandwidth remaining percent 30
    random-detect dscp-based
    set dscp af31
  class STREAMING-VIDEO
    bandwidth remaining percent 10
    random-detect dscp-based
    set dscp af31
  class NET-CTRL-MGMT
    bandwidth remaining percent 5
    set dscp cs6
  class CALL-SIGNALING
    bandwidth remaining percent 4
    set dscp af21
  class CRITICAL-DATA
    bandwidth remaining percent 25
    random-detect dscp-based
    set dscp af21
  class SCAVENGER
    bandwidth remaining percent 1
    set dscp default
  class VOICE
    priority level 1
    police cir percent 10
    set dscp ef
  class class-default
    bandwidth remaining percent 25
    random-detect
    set dscp default
```

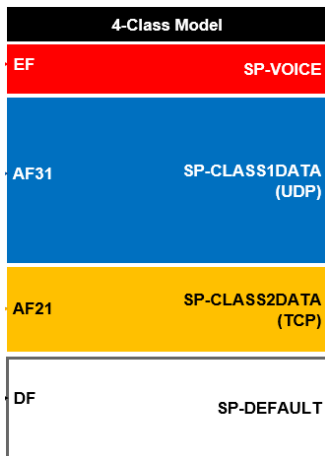
Branch Router:

```
policy-map POLICY-TRANSPORT-1
  class class-default
    shape average 10 Mbps
    service-policy EDGE-QUEUEING
```

```
interface GigabitEthernet0/0
  bandwidth 10000
  service-policy output POLICY-TRANSPORT-1
```

4-Class SP QoS Model Configuration

Tunneled Traffic



```
policy-map EDGE-QUEUING
  class INTERACTIVE-VIDEO
    bandwidth remaining percent 30
    random-detect dscp-based
    set dscp tunnel af31
  class STREAMING-VIDEO
    bandwidth remaining percent 10
    random-detect dscp-based
    set dscp tunnel af31
  class NET-CTRL-MGMT
    bandwidth remaining percent 5
    set dscp tunnel cs6
  class CALL-SIGNALING
    bandwidth remaining percent 4
    set dscp tunnel af21
  class CRITICAL-DATA
    bandwidth remaining percent 25
    random-detect dscp-based
    set dscp tunnel af21
  class SCAVENGER
    bandwidth remaining percent 1
    set dscp tunnel default
  class VOICE
    priority level 1
    police cir percent 10
    set dscp tunnel ef
  class class-default
    bandwidth remaining percent 25
    random-detect
    set dscp tunnel default
```

Hub Router:

```
policy-map GROUP-10MBPS-POLICY
  class class-default
    shape average 10 Mbps
    bandwidth remaining ratio 10
    service-policy EDGE-QUEUING
```

```
interface Tunnel10
  bandwidth <service-rate>
  nhrp map group GROUP-10MBPS service-policy
  output GROUP-10MBPS-POLICY
```

Branch Router:

```
interface GigabitEthernet0/0
  bandwidth 10000
  service-policy output POLICY-TRANSPORT-1
!
interface Tunnel10
  bandwidth 10000
  nhrp group GROUP-10MBPS
  tunnel source GigabitEthernet0/0
  tunnel vrf TRANSPORT-1
```

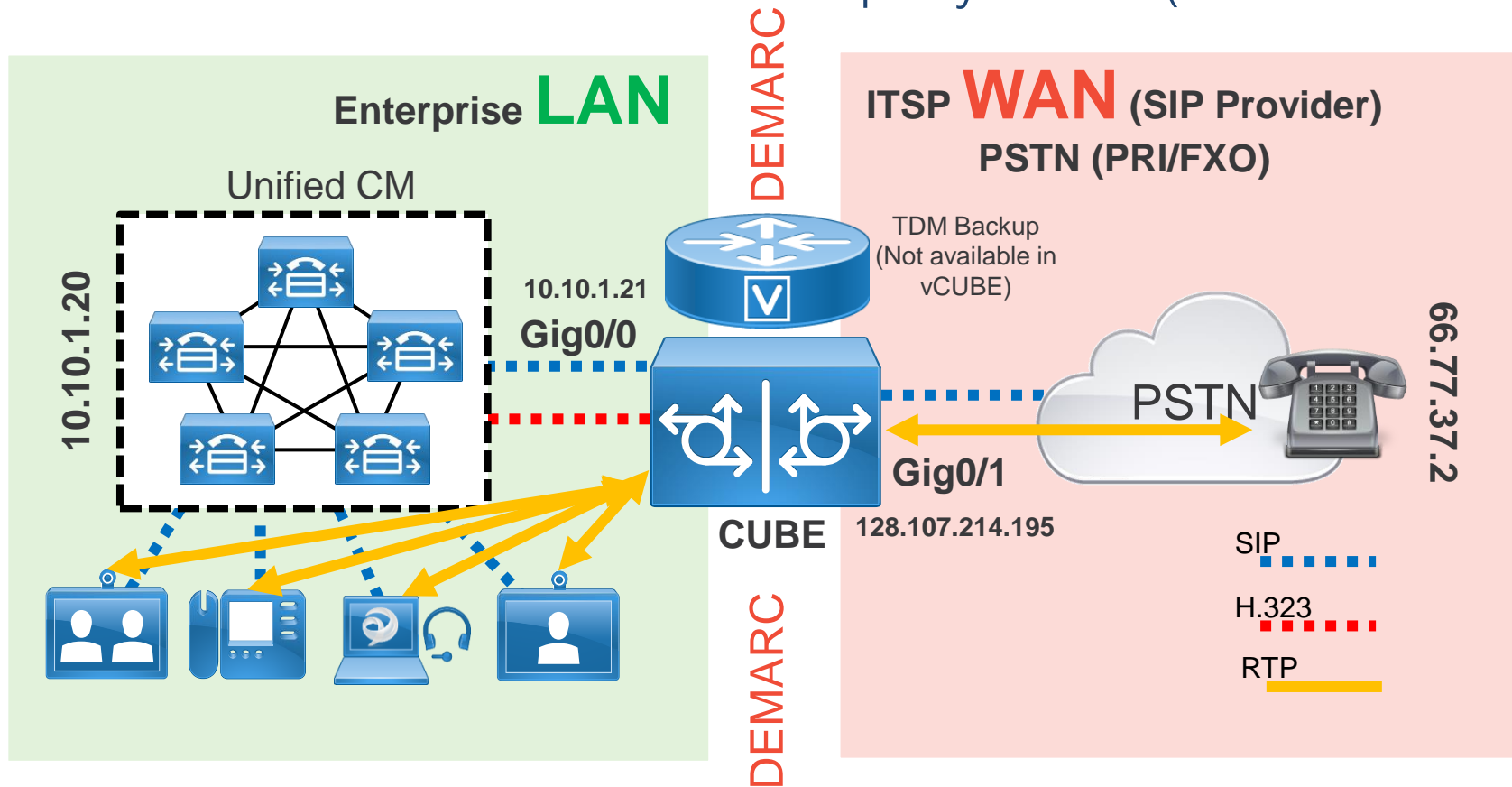
Agenda

- WAN / VPN QoS Design
- **Integrated Voice. CUBE**

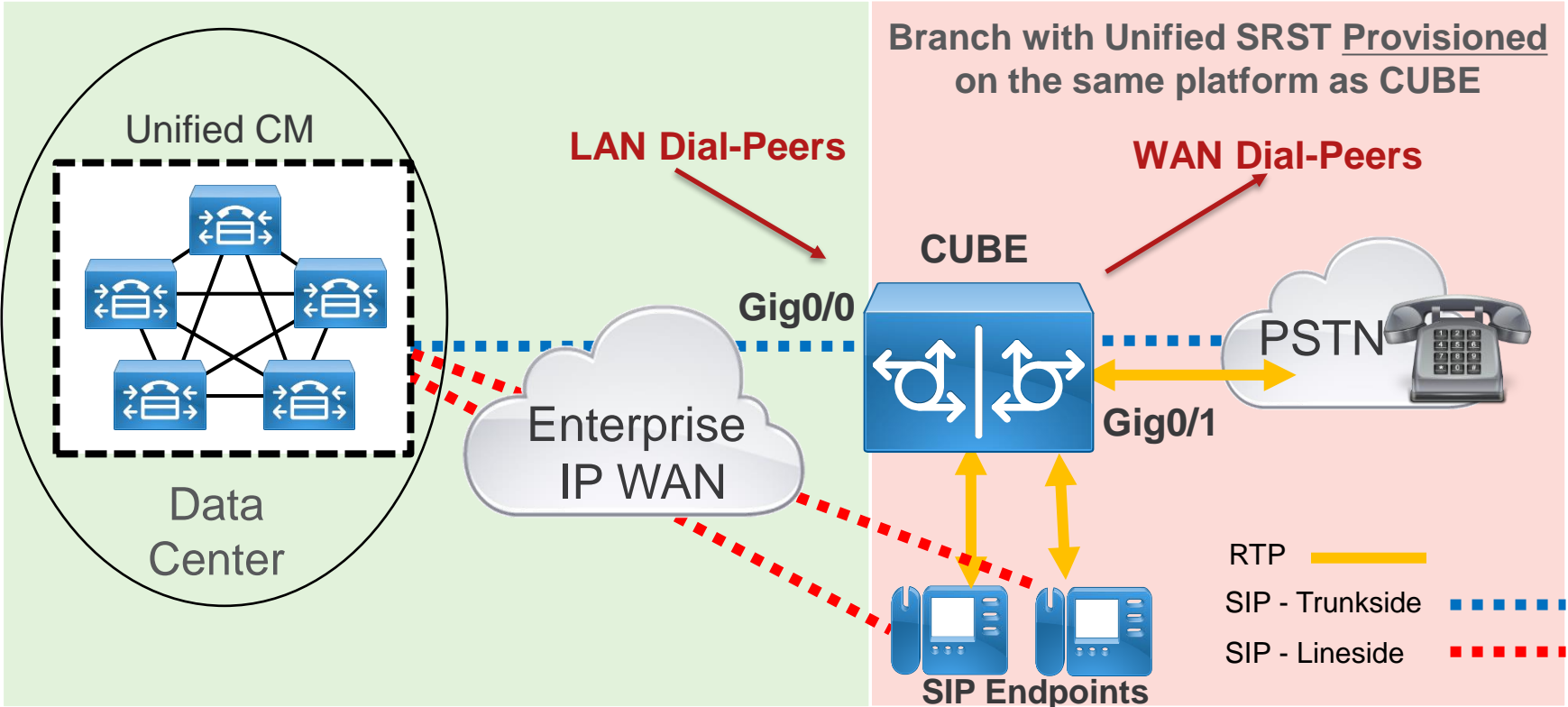
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CUBE Overview and Deployments

On-Prem Collaboration Deployment (CUBE-T-STD)



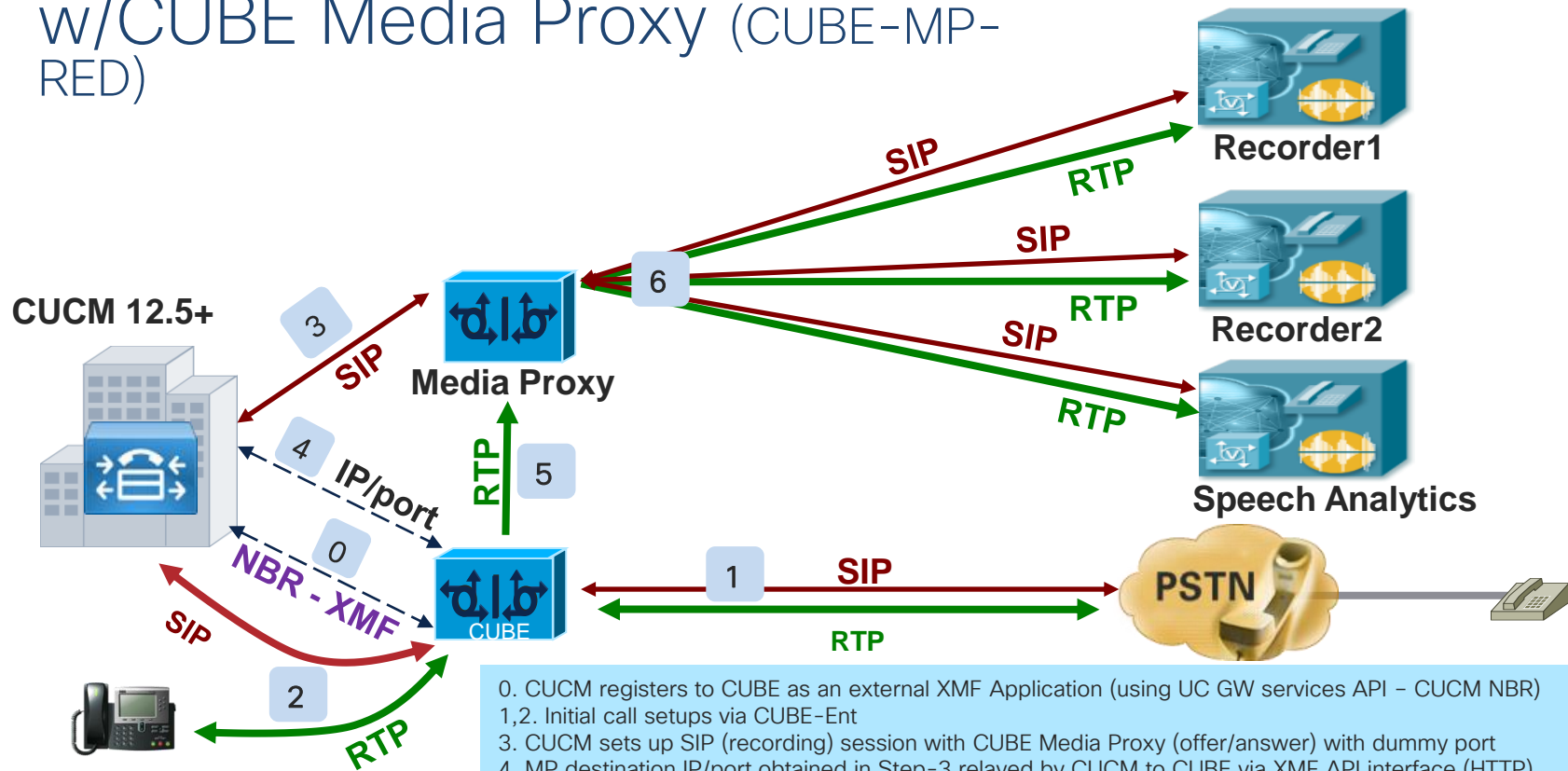
Branch CUBE Deployment with SRST Provisioned (CUBE-T-STD)



Enterprise **LAN**

ITSP **WAN** (SIP Provider)

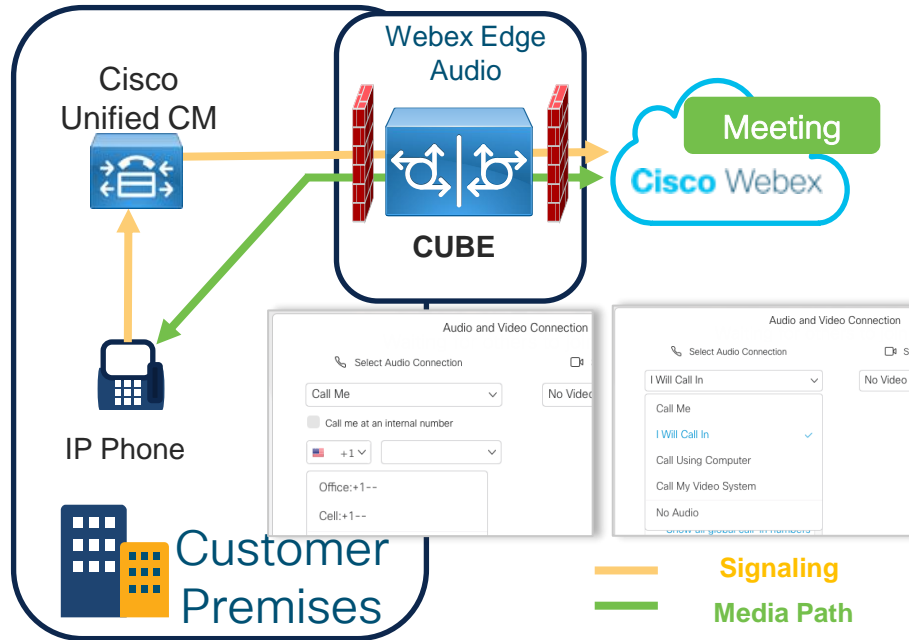
Enabling compliance recording w/CUBE Media Proxy (CUBE-MP-RED)



0. CUCM registers to CUBE as an external XMF Application (using UC GW services API - CUCM NBR)
 - 1,2. Initial call setups via CUBE-Ent
 3. CUCM sets up SIP (recording) session with CUBE Media Proxy (offer/answer) with dummy port
 4. MP destination IP/port obtained in Step-3 relayed by CUCM to CUBE via XMF API interface (HTTP)
 5. CUBE-Ent starts to fork media streams to the MP (target ip/port received in Step-4). MP accepts RTP because of Media latching in the inbound leg from CUCM
 6. MP sets up SIP recording sessions with the 3 Recorders for multi-fork.
- The ingress media stream from CUBE-Ent is then multi-forked by MP towards the 3 recorders simultaneously using the destination ip/ports as negotiated in the SIP offer/answer b/w MP and the Recorders.

Deploying Cisco Webex Edge Audio w/CUBE

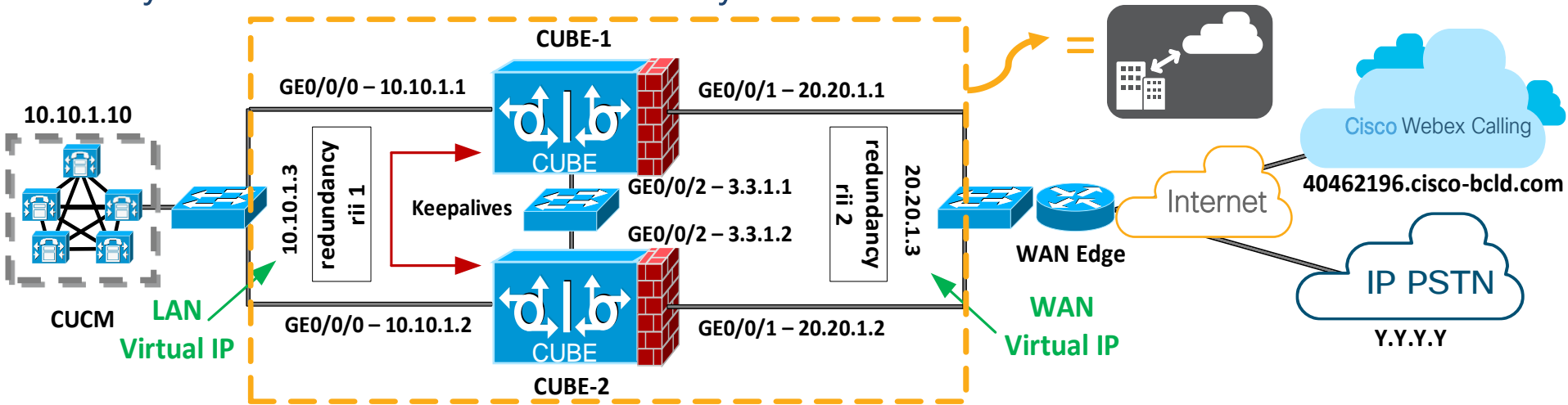
High level overview



1. On-premises telephone dials the Webex meeting number or gets a call back from the Webex meeting to get connected by audio into the meeting.
2. Signaling is routed via the on-premises call control device (Unified CM) through the CUBE to Webex Meetings audio service.
3. Audio media (the sound) is routed from the Webex meeting to CUBE and then to the on-premises phone for callback and the reverse for call in.

CUBE High Availability as Local Gateway

Layer 2 box-to-box redundancy

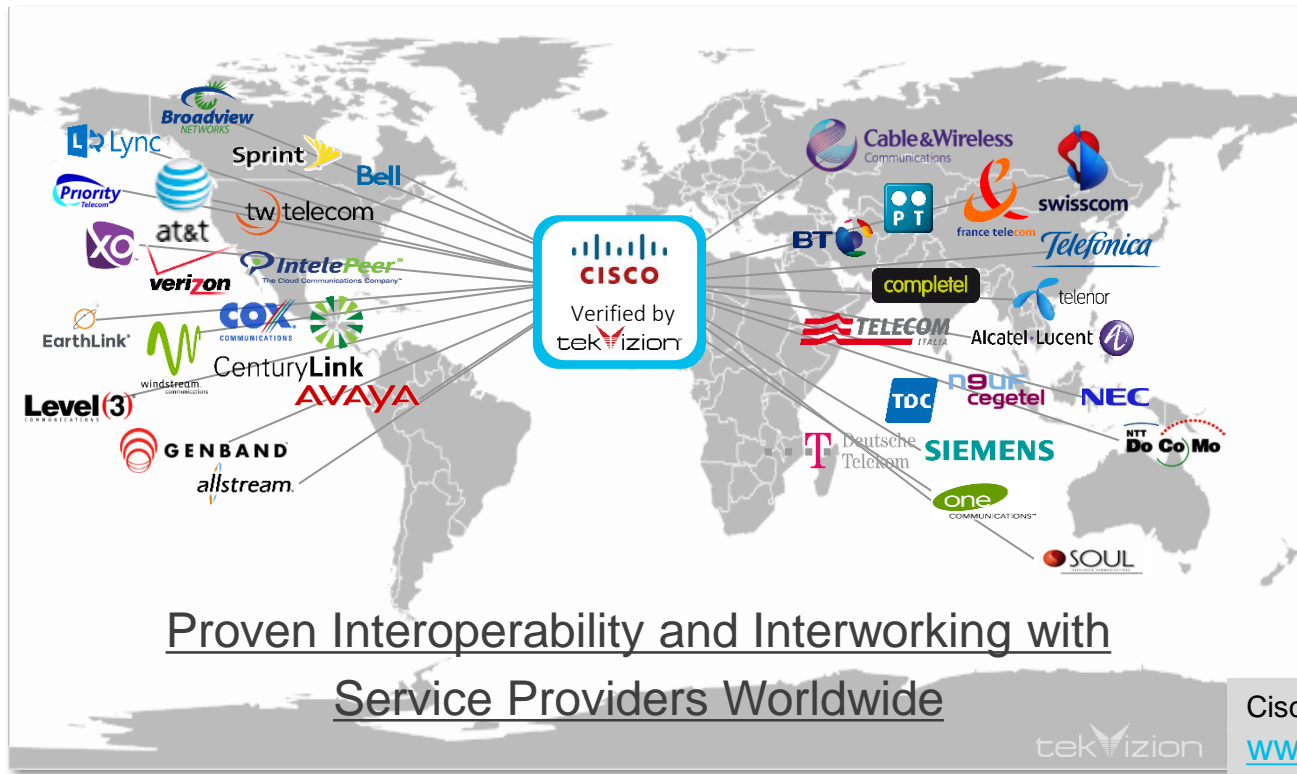


- LGW HA solution with layer 2 box-to-box redundancy for call preservation
 - CUBE HA Active/standby model using virtual IP addresses
 - Applicable to ISR 4K and vCUBE only
 - Acts as a **single Local Gateway** from Webex Calling point of view
- Support for Webex Calling deployments available from **IOS-XE 16.12.2**
- LGW HA cannot have TDM or analog interfaces co-located

CUBE Interoperability Portal for application note



- Validated with Service Providers World-Wide
- Independently Tested with 3-Party PBXs in tekVizion Labs
- Standards based



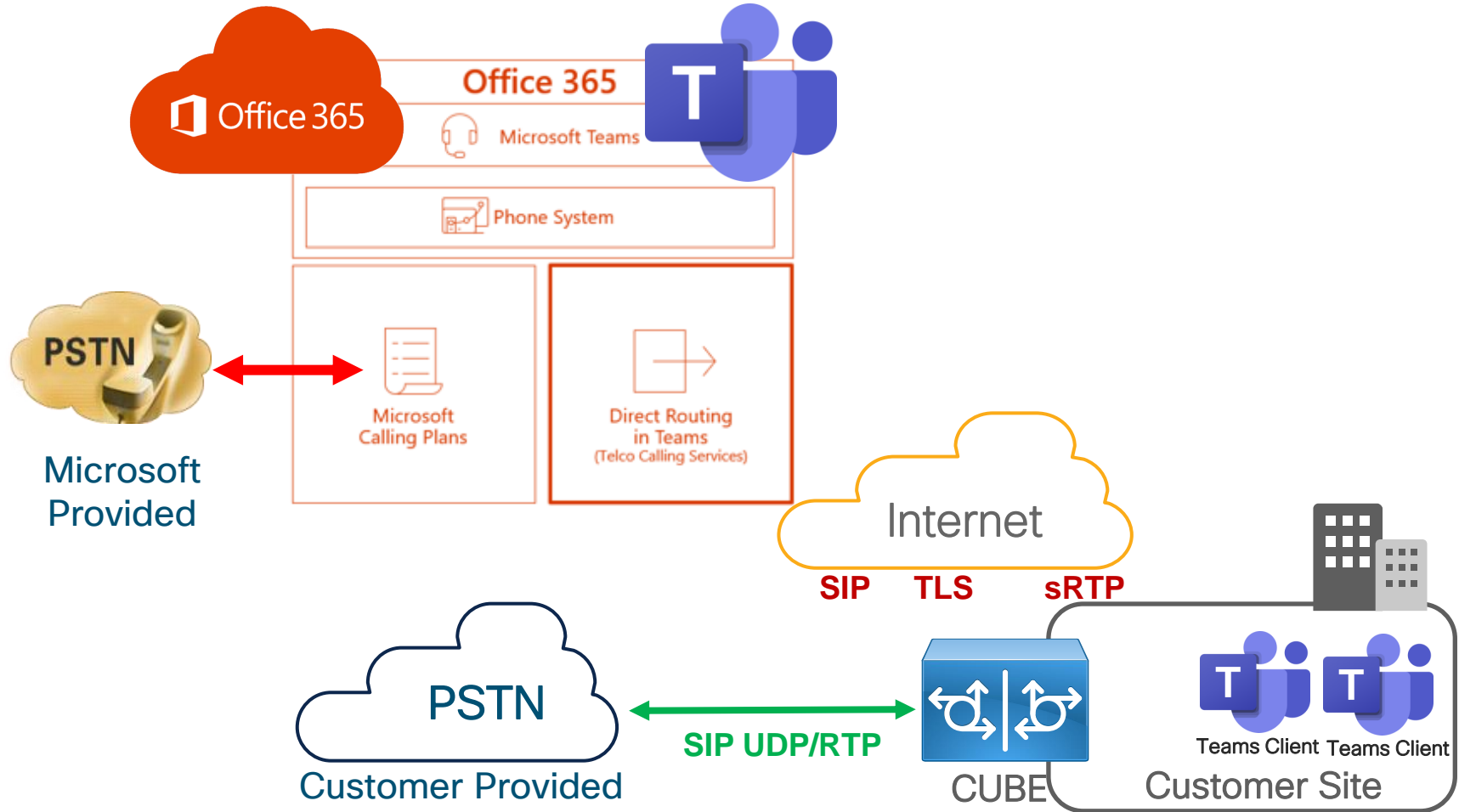
Proven Interoperability and Interworking with
Service Providers Worldwide

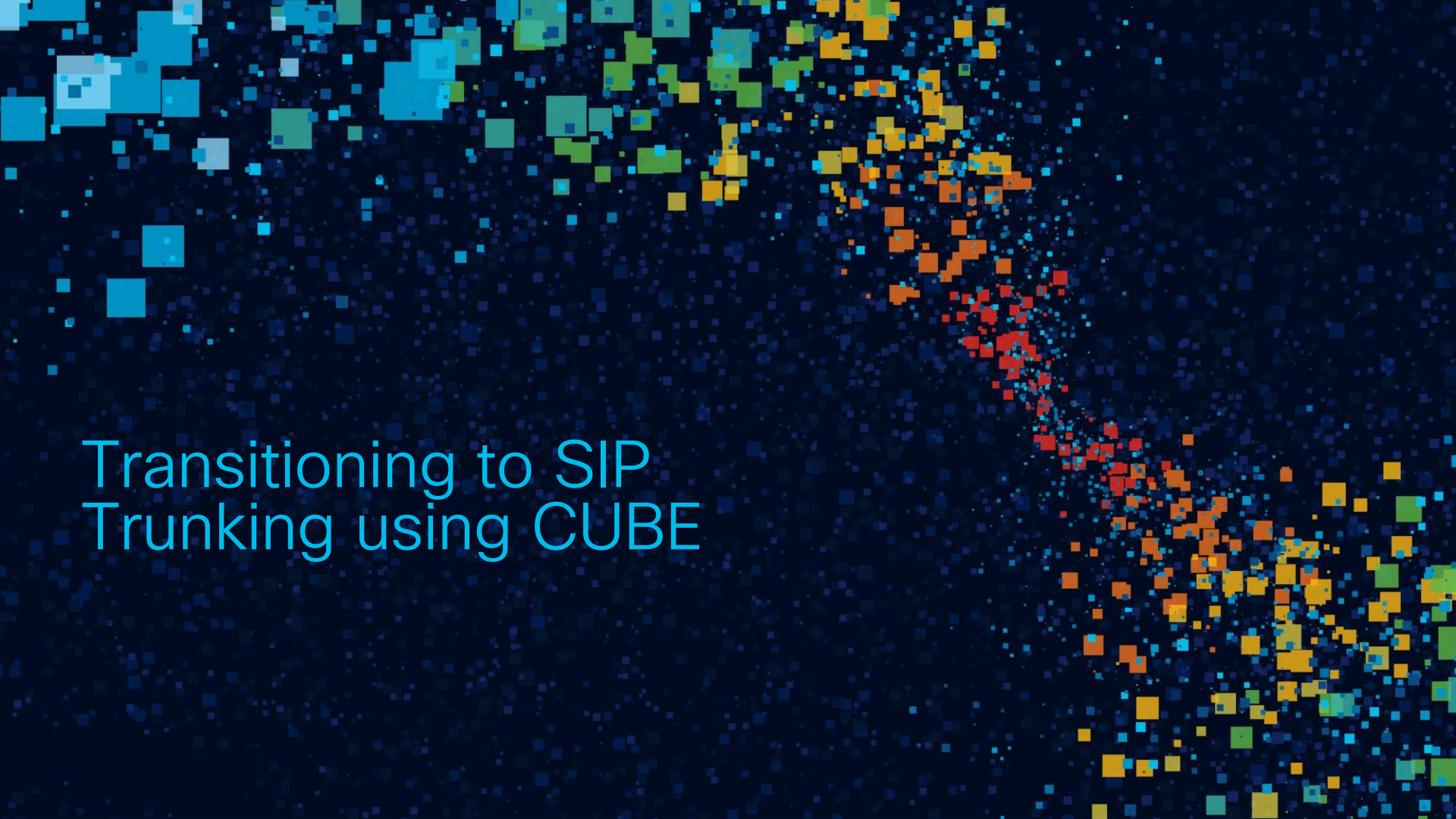


Cisco Interoperability Portal:

www.cisco.com/go/interoperability

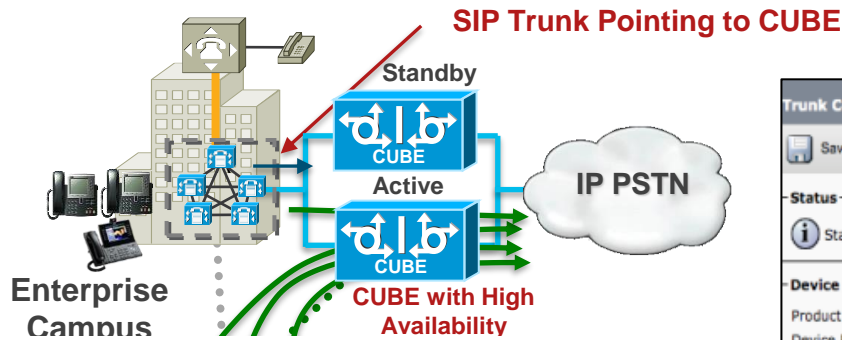
Microsoft Teams Direct Routing – Solution Overview



The background is a dark blue field filled with numerous small, semi-transparent squares and dots. These elements are scattered across the frame, with a higher concentration of larger squares in the upper left and lower right corners. The colors of these elements include various shades of blue, green, yellow, orange, and red, creating a vibrant, pixelated effect.

Transitioning to SIP Trunking using CUBE

Step 1: Configure IP PBX to route calls to the edge SBC



- Configure CUCM to route all PSTN calls (central and branch) to CUBE (Gig0/0 in our slides) via a SIP trunk
- Make sure all different patterns of calls – local, long distance, international, emergency, informational etc.. are pointing to CUBE

Trunk Configuration

Save X Delete Reset + Add New

Status

Status: Ready

Device Information

Product: SIP Trunk
Device Protocol: SIP
Device Name*: CUBE_SIP_Trunk
Description: SIP Trunk to CUBE
Device Pool*: Default
Common Device Configuration: < None >
Call Classification*: Use System Default
Media Resource Group List: < None >
Location*: Hub_None
AAR Group: < None >
Packet Capture Mode*: None
Packet Capture Duration: 0

☐ Media Termination Point Required
☒ Retry Video Call as Audio
☐ Transmit UTF-8 for Calling Party Name
☐ Unattended Port

Step 2: Get details from SIP Trunk provider

Item	SIP Trunk service provider requirement	Sample Response
1	SIP Trunk IP Address (Destination IP Address for INVITES)	66.77.37.2 or DNS
2	SIP Trunk Port number (Destination port number for INVITES)	5060
3	SIP Trunk Transport Layer (UDP or TCP)	UDP
4	Codecs supported	G711, G729
5	Fax protocol support	T.38
6	DTMF signaling mechanism	RFC2833
7	Does the provider require SDP information in initial INVITE (Early offer required)	Yes
8	SBC's external IP address that is required for the SP to accept/authenticate calls (Source IP Address for INVITES)	128.107.214.195
9	Does SP require SIP Trunk registration for each DID? If yes, what is the username & password	No
10	Does SP require Digest Authentication?	408-944-7700

Step 3: Enable CUBE Application on Cisco routers

1. Enable CUBE Application

voice service voip

mode border-element

→ Enables CUBE, *capacity* keyword has been deprecated.

allow-connections sip to sip

→ By default IOS/IOS-XE voice devices do not allow an incoming VoIP leg to go out as VoIP

2. Configure any other global settings to meet SP's requirements

voice service voip

media bulk-stats →

To increment Rx/Tx counters on IOS-XE based platforms. W/O this CLI, it will show 0/0 (CPU intensive CLI)

sip

early-offer forced

3. Create a trusted list of IP addresses to prevent toll-fraud

voice service voip

ip address trusted list →

Applications initiating signaling towards CUBE, e.g. CUCM, CVP,

ipv4 66.77.37.2 ! ITSP SIP Trunk

Service Provider's SBC. IP Addresses from dial-peers with "session target ip" or Server Group are trusted by default and need not be populated here

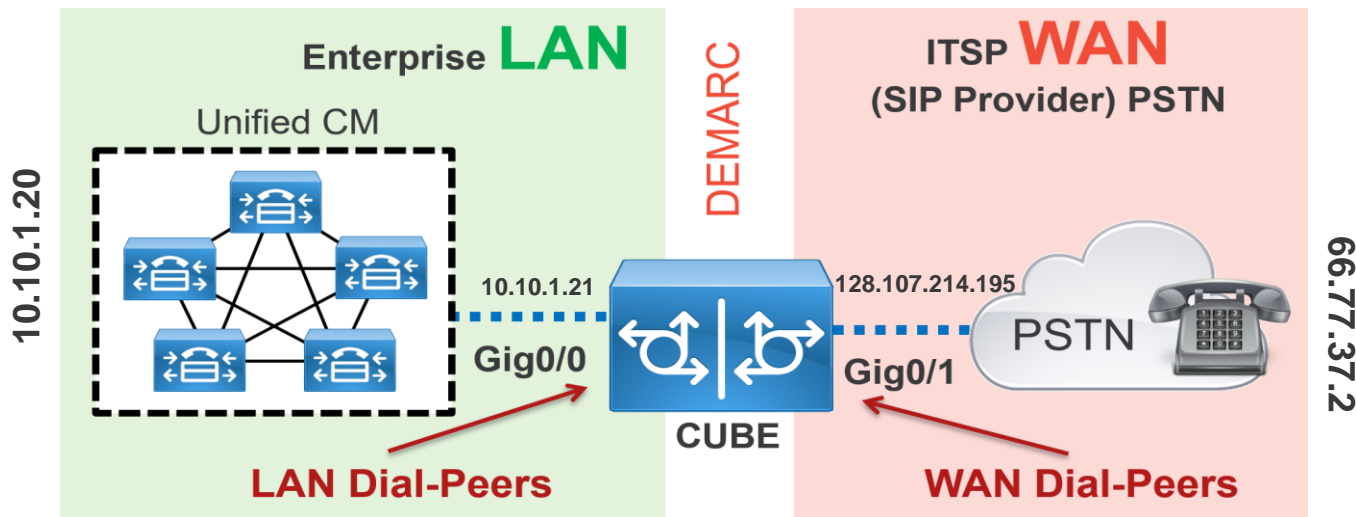
ipv4 10.10.1.20 ! CUCM

sip

silent-discard untrusted →

Default configuration starting XE 3.10.1 /15.3(3)M1 to mitigate TDoS Attack

Step 4: Configure Call routing on CUBE



- Dial-Peer – “static routing” table mapping phone numbers to interfaces or IP addresses
- LAN Dial-Peers – Dial-peers that are facing towards the IP PBX for sending and receiving call legs to and from the PBX. Always bind LAN interface(s) on CUBE to LAN dial-peers, ensuring SIP/RTP is sourced from the intended LAN interfaces(s)
- WAN Dial-Peers – Dial-peers that are facing towards the SIP Trunk provider for sending and receiving call legs to and from the ITSP. Always bind CUBE’s WAN interface(s) to WAN dial-peer(s).

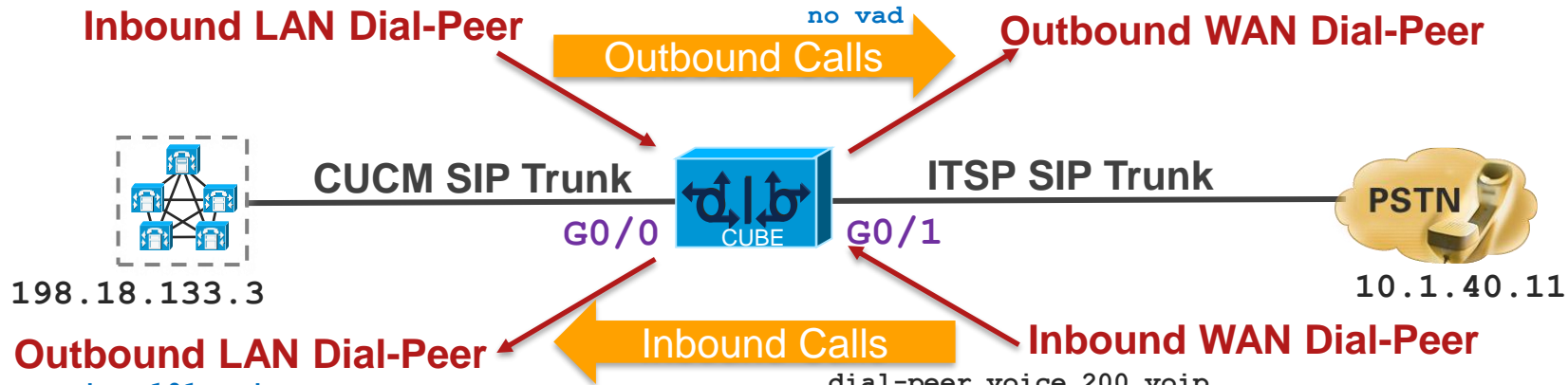
The background is a dark blue field filled with numerous small, semi-transparent squares and dots. These elements are scattered across the frame, with a higher concentration of larger squares in the upper left and lower right corners. The colors of these elements include various shades of blue, cyan, green, yellow, orange, and red, creating a vibrant, pixelated effect.

CUBE Dial-Peers

Advanced Call Routing

```
dial-peer voice 100 voip
description *Inbound LAN dial-peer. From CUCM to CUBE*
session protocol sipv2
incoming called-number 8T
voice-class sip bind control source-interface Gig0/0
voice-class sip bind media source-interface Gig0/0
dtmf-relay rtp-nte
codec g711ulaw
no vad
```

```
dial-peer voice 201 voip
description *Outbound WAN dial-peer. From CUBE to SP*
destination-pattern 81[2-9]..[2-9].....$
session protocol sipv2
session target ipv4:10.1.40.11
session transport udp
voice-class sip bind control source-interface Gig0/1
voice-class sip bind media source-interface Gig0/1
dtmf-relay rtp-nte
codec g711ulaw
no vad
```



```
dial-peer voice 101 voip
description *Outbound LAN dial-peer. From CUBE to CUCM*
translation-profile outgoing CUBE_to_CUCM
destination-pattern +1408944....$
session protocol sipv2
session target ipv4:198.18.133.3
voice-class sip bind control source-interface Gig0/0
voice-class sip bind media source-interface Gig0/0
dtmf-relay rtp-nte
codec g711ulaw
no vad
```

```
dial-peer voice 200 voip
description *Inbound WAN dial-peer. From Provider to CUBE*
session protocol sipv2
incoming uri via 200
voice-class sip bind control source-interface Gig0/1
voice-class sip bind media source-interface Gig0/1
dtmf-relay rtp-nte
codec g711ulaw
no vad

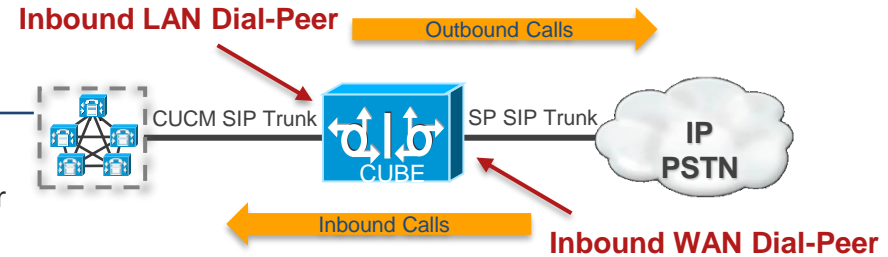
voice class uri 200 sip
host ipv4:10.1.40.11
```

Understanding Inbound Dial-Peer Matching Techniques

Priority

Dial-Peer Matching Rules for Inbound URI in SIP Calls

Match Order	Cisco IOS Command	Incoming Call Parameter
1	incoming uri via	Via URI
2	incoming uri request	Request-URI
3	incoming uri to	To URI
4	incoming uri from	From URI
5	incoming called-number	Called number
6	incoming called e164-pattern-map	Called Pattern-Map-Number
7	incoming calling e164-pattern-map	Calling Pattern-Map-Number
8	answer-address	Calling number
9	destination-pattern	Calling number
10	carrier-id source	Carrier-ID associated with the call



Received:

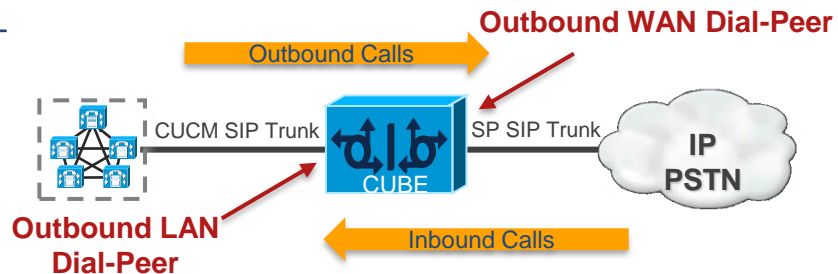
```
INVITE sip:654321@10.2.1.1 SIP/2.0
Via: SIP/2.0/UDP 10.1.1.1:5060;x-route-
tag="cid:orange@10.1.1.1";branch=z9hG4bK-23955-
1-0
From: "555" <sip:555@10.1.1.1:5060>;tag=1
To: ABC <sip:654321@10.2.1.1:5060>
Call-ID: 1-23955@10.1.1.1
CSeq: 1 INVITE
Contact: sip:555@10.1.1.1:5060
Supported: timer
Max-Forwards: 70
Subject: BRKUCC-2934 Session
Content-Type: application/sdp
Content-Length: 226
```

Outbound Dial-Peer Matching Criteria Summary

Priority

Dial-Peer Matching Rules for Outbound URI in SIP Calls

Match Order	Cisco IOS Command	Call Parameter
1	destination dpkg <dpkg-tag> (configured on inbound dial-peer)	Dial-peer Group Dial-peer
2	destination uri-from <uri-tag>	From URI
3	destination uri-to <uri-tag>	To URI
4	destination uri-via <uri-tag>	Via URI
5	destination uri-diversion <uri-tag>	Diversion URI
6	destination uri-referred-by <uri-tag>	URI-Referred-by URI
7	destination route-string <route-string-tag>	ILS Route String
8	destination uri <uri-tag> AND carrier-id target <string>	URI and Carrier-ID
9	destination-pattern <number-string> AND carrier-id target <string>	Called Number and Carrier-ID
10	destination uri <uri-tag>	URI
11	destination-pattern <DNIS-number>	Called Number
12	destination e164-pattern-map <pattern-map-number>	Called E164 Pattern Map
13	dnis-map <dnis-map-number>	Called DNIS Map
14	destination calling e164-pattern-map <pattern-map-number>	Calling E164 Pattern Map



Received:

```
INVITE sip:654321@10.2.1.1 SIP/2.0
Via: SIP/2.0/UDP 10.1.1.1:5060;x-route-tag="cid:orange@10.1.1.1";branch=z9hG4bK-23955-1-0
From: "555" <sip:555@10.1.1.1:5060>;tag=1
To: ABC <sip:654321@10.2.1.1:5060>
Call-ID: 1-23955@10.1.1.1
CSeq: 1 INVITE
Contact: sip:555@10.1.1.1:5060
Supported: timer
Max-Forwards: 70
Subject: BRKUCC-2934 Session
Content-Type: application/sdp
Content-Length: 226
.....
```

Destination Server Group

- Supports multiple destinations (session targets) be defined in a group and applied to a single outbound dial-peer
- Once an outbound dial-peer is selected to route an outgoing call, multiple destinations within a server group will be sorted in either round robin or preference [**default**] order
- This reduces the need to configure multiple dial-peers with the same capabilities but different destinations. E.g. Multiple subscribers in a cluster

```
voice class server-group 1
  hunt-scheme {preference | round-robin}
  ipv4 1.1.1.1 preference 5
  ipv4 2.2.2.2
  ipv4 3.3.3.3 port 5065 preference 3
  ipv6 2010:AB8:0:2::1 port 5065 preference 3
  ipv6 2010:AB8:0:2::2
```

* DNS target not supported in server group

```
dial-peer voice 100 voip
  description Outbound DP
  destination-pattern 1234
  session protocol sipv2
  codec g711ulaw
  dtmf-relay rtp-nte
  session server-group 1
```


Multiple Number Patterns Under Same Incoming/Outgoing Dial-peer

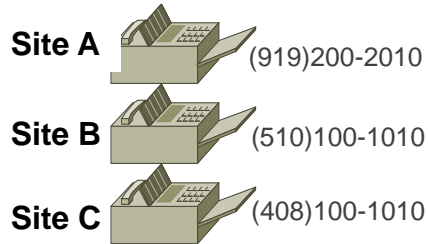
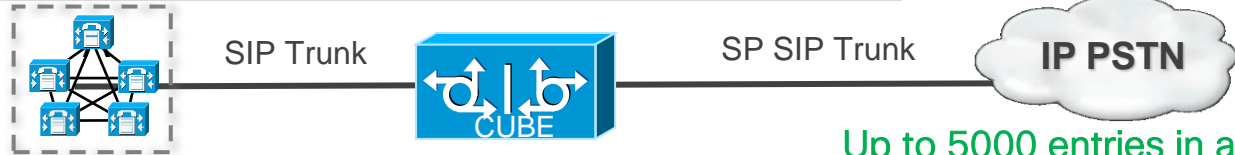


G729 Sites

```
voice class e164-pattern-map 300
  e164 200.
  e164 510100100.
  e164 408100100.

dial-peer voice 1 voip
  description Inbound DP via Calling
  incoming calling e164-pattern-map 300
  codec g729r8
```

Up to 1000 entries
in a pattern map



G711 Sites

```
voice class e164-pattern-map 400
  url flash:e164-pattern-map.cfg

dial-peer voice 2 voip
  description Outbound DP via Called
  destination e164-pattern-map 400
  codec g711ulaw
```

Up to 5000 entries in a text file

! This is an example of the contents
of E164 patterns text file stored
in flash:**e164-pattern-map.cfg**

```
9192002010
5101001010
4081001010
<blank line>
```

Destination Dial-peer Group

```
voice class dpg 10000
  description Voice Class DPG for SJ
  dial-peer 1001 preference 1
  dial-peer 1002 preference 2
  dial-peer 1003
```

```
!
dial-peer voice 100 voip
  description Inbound DP
  incoming called-number 1341
  destination dpg 10000
```

Received:

INVITE sip:1341@CUBE-IP-ADDRESS:5060

1. Incoming Dial-peer is first
Sent: matched

INVITE sip:1341@10.1.1.3:5060

3. Outbound
DP is
selected

```
dial-peer voice 1001 voip
  destination-pattern BAD
  session protocol sipv2
  session target ipv4:10.1.1.1
```

!

```
dial-peer voice 1002 voip
  destination-pattern BAD.BAD
  session protocol sipv2
  session target ipv4:10.1.1.2
```

!

```
dial-peer voice 1003 voip
  destination-pattern BAD.BAD.BAD
  session protocol sipv2
  session target ipv4:10.1.1.3
```

2. Now the DPG associated with
the INBOUND DP is selected

Call Admission Control (CAC)

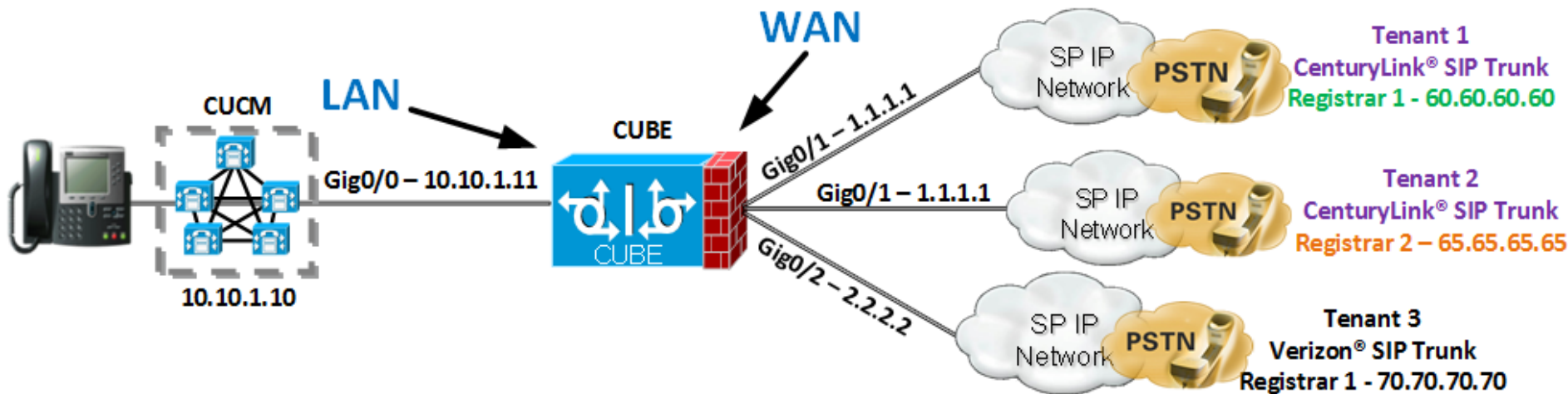
- Call processing capacity for any CUBE instance will be influenced by several considerations, including software version, features configured and the platform itself
- To ensure that calls continue to be processed reliably, configure Call Admission Control as follows to reject calls when use of system resources exceeds 80%. Refer to the [CUBE Configuration Guide](#) for further details

```
enable
conf t
  call threshold global cpu-avg low 75 high 80
  call threshold global total-mem low 75 high 80
  call treatment on
end
```

- `show call active total-calls` lists the total number of concurrent calls on a CUBE platform

Multi-Tenancy

Multiple Tenants on CUBE



- Every Registrar/User Agent/ITSP connected to CUBE can be considered a Tenant to CUBE
- Allows specific global configurations (CLI under sip-ua) for multiple tenants such as specific SIP Bind for REGISTER messages
- Allows differentiated services for different tenants

Configuring Voice Class Tenant

- Configure voice class tenant

```
voice class tenant 1
```

```
  registrar 1 ipv4:10.64.86.35:9052 expires 3600
  credentials username aaaa password 7 06070E204D realm aaaa.com
  credentials number bbbb username bbbb password 7 110B1B0715 realm bbbb.com
  bind control source-interface GigabitEthernet0/0
  bind media source-interface GigabitEthernet0/0
  copy-list 1
  outbound-proxy ipv4:10.64.86.35:9055
  early-offer forced
```

Add new voice class tenant

- Apply tenant to the desired dial-peer

```
dial-peer voice 1 voip
  destination-pattern 111
  session protocol sipv2
  session target ipv4:10.64.86.35:9051
  session transport udp
  voice-class sip tenant 1
```

Configuration Preference Order

1. dial-peer (overrides tenant and global configs)
2. tenant (overrides the global config)
3. global (voice service voip OR sip-ua)

Apply Tenant to a Dial-peer

The background is a dark blue field filled with numerous small, semi-transparent squares and dots. These elements are scattered across the frame, with a higher concentration in the upper left and lower right areas. The colors of these elements include light blue, teal, yellow, orange, and red, creating a dynamic, pixelated effect.

External/PSTN Call Recording

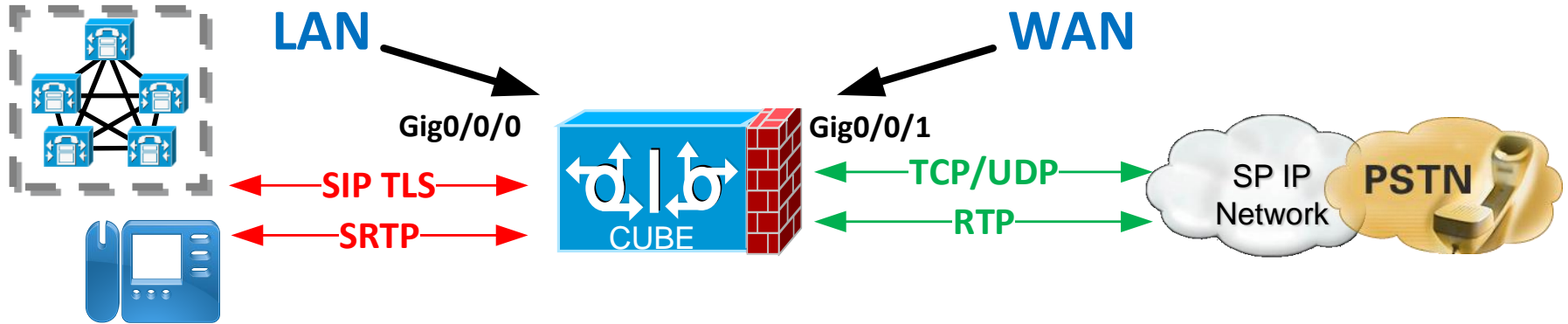
External/PSTN Call Recording Options

- CUBE Controlled (Dial-peer based SIPREC)
 - SIPREC based, CUBE sends metadata in XML format
 - Dial-peer controlled, IP-PBX independent
 - Source of recorded media (RTP only) is always CUBE (External calls only).
 - Records both audio and video calls and supported with CUBE HA
- CUCM NBR (Network Based Recording)
 - CUCM Controlled & triggered, requires UC Services API be enabled on CUBE
 - Audio calls only
 - Source of Recorded Media can be CUBE (Gateway Preferred) or Phone based (BiB)

The background is a dark blue field filled with numerous small, semi-transparent squares and dots. These elements are scattered across the frame, with a higher concentration on the left side. The colors of these elements include various shades of blue, green, yellow, orange, and red, creating a vibrant, pixelated effect.

Securing Collab deployments with CUBE

Secure SIP Trunks with CUBE



- Interworking between all three transport types is supported : UDP/TCP/TLS
- IOS-XE based platforms do not require DSPs for SRTP-RTP interworking
- TLS Exclusivity can be configured with “transport tcp tls v1.2”
- NGE Crypto supported for SRTP-SRTP (IOS-XE 16.5.2) [Crypto A – Crypto B], SRTP-RTP, SRTP pass-thru

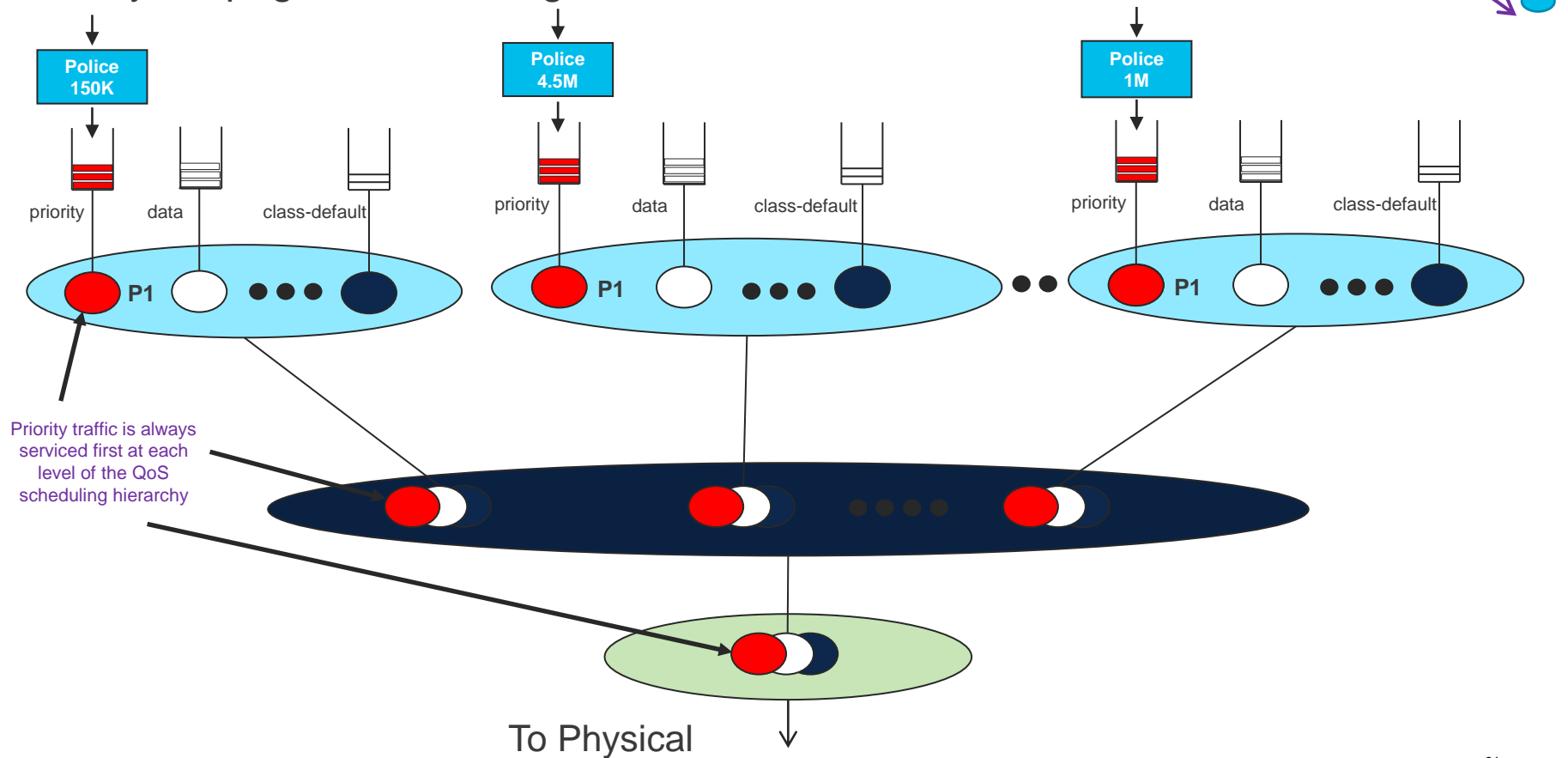
Thank you



Appendix A QoS Design

Aggregate Priority Load

Priority Propagation / Passing Lanes



Aggregate Priority Load

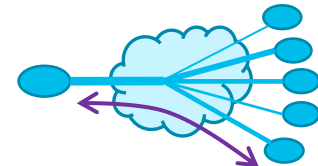
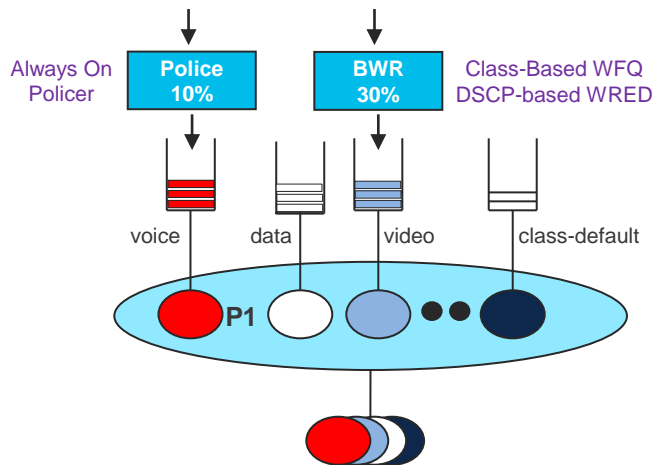
Conclusion

- For Voice, use an **Always On policer**, rather than a Conditional policer

```
class VOICE
  priority level 1
  police cir percent 10
```

- For Video, use a **Bandwidth Remaining Percent (BWR)** queue with **DSCP-based WRED**, rather than a level 2 Priority queue

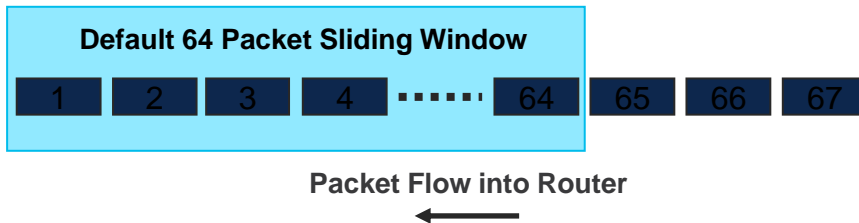
```
class INTERACTIVE-VIDEO
  bandwidth remaining percent 30
  random-detect dscp-based
```



IPsec Anti-Replay

Message Integrity

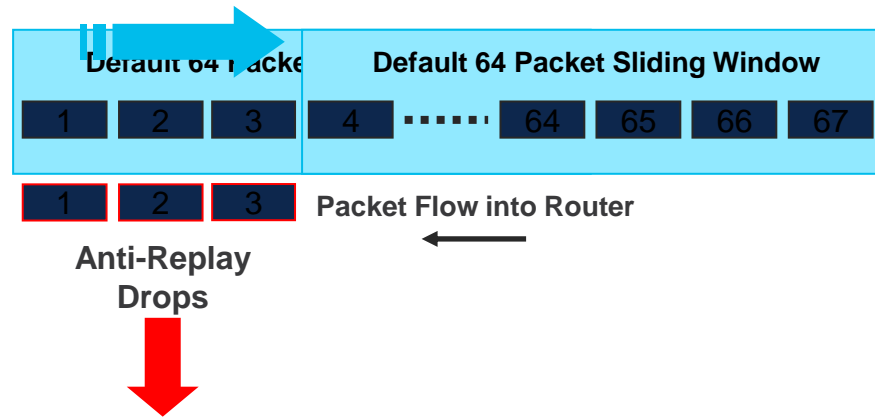
- Designed to identify packet capture/replay by 3rd party – Message Integrity
- Sender assigns sequence number per Security Association (SA) to encrypted packets
- Receiver maintains 64 packet sliding window by default



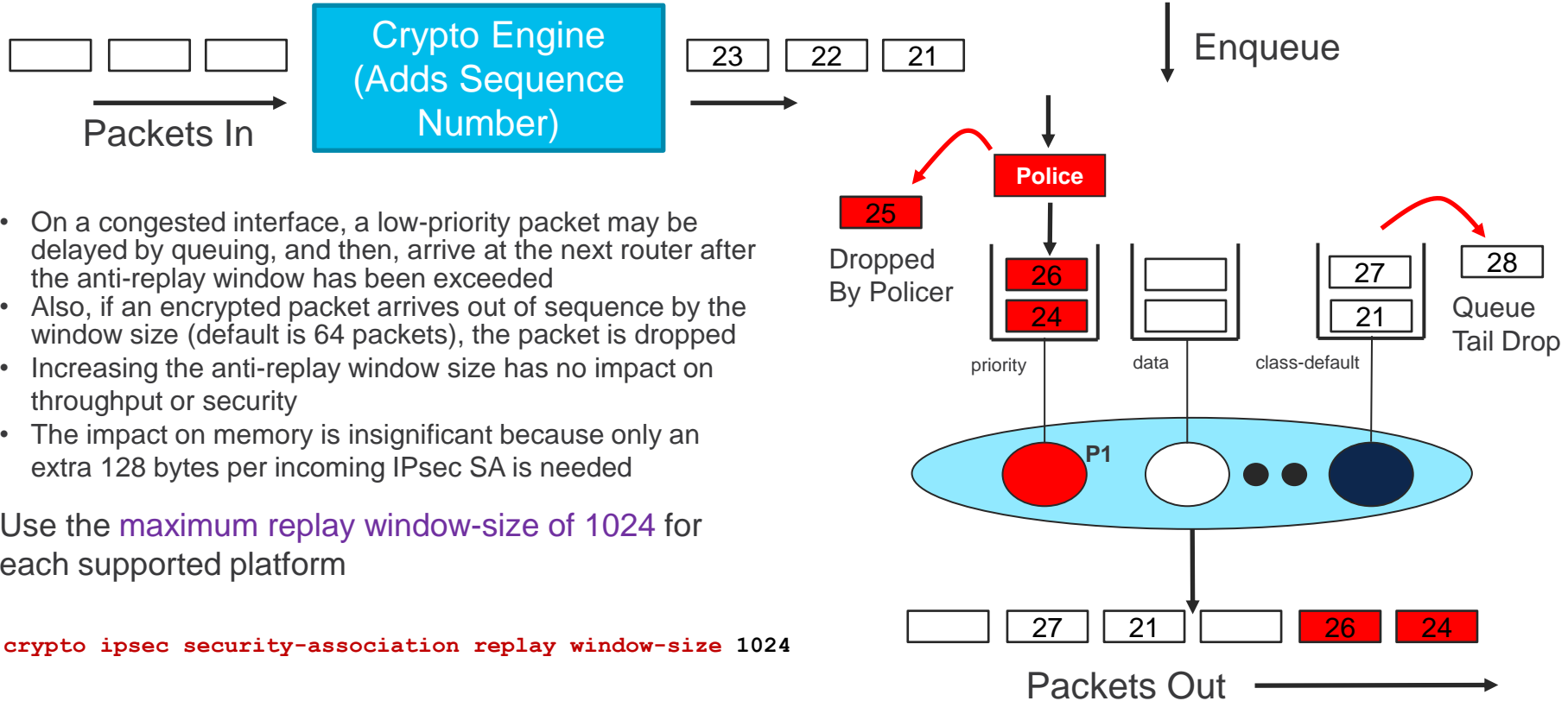
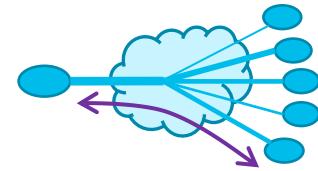
IPsec Anti-Replay

Message Integrity

- Designed to identify packet capture/replay by 3rd party – Message Integrity
- Sender assigns sequence number per Security Association (SA) to encrypted packets
- Receiver maintains 64 packet sliding window by default
- Window moves right to include higher sequence numbers
- Window marks packets as received or not
- Packets to the left of the window are dropped



IPsec Anti-Replay and QoS



- On a congested interface, a low-priority packet may be delayed by queuing, and then, arrive at the next router after the anti-replay window has been exceeded
- Also, if an encrypted packet arrives out of sequence by the window size (default is 64 packets), the packet is dropped
- Increasing the anti-replay window size has no impact on throughput or security
- The impact on memory is insignificant because only an extra 128 bytes per incoming IPsec SA is needed

Use the **maximum replay window-size of 1024** for each supported platform

```
crypto ipsec security-association replay window-size 1024
```


QoS Tools Review: Queuing and Dropping Tools

Bandwidth Percent vs Bandwidth Remaining Percent

Bandwidth Percent specifies bandwidth allocation as a percentage of the value entered in the bandwidth command on the interface

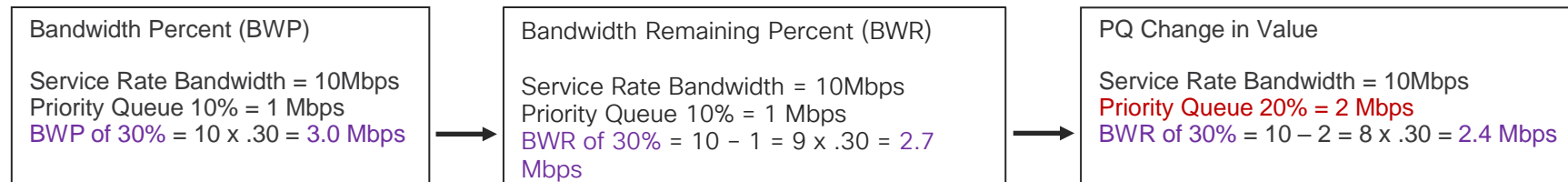
- Bandwidth percentages have to take into account priority percent values
- They have to be adjusted when priority bandwidth values are changed

Bandwidth Remaining Percent specifies bandwidth allocation as a percentage of the bandwidth value that has **not** been allocated to priority classes

- Bandwidth remaining percentages must equal 100%
- The bandwidth automatically adjusts when priority bandwidth values are changed

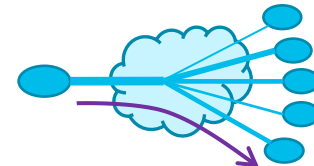
The two features cannot be used in the same policy map

Examples:



Bandwidth Remaining Ratio

Details

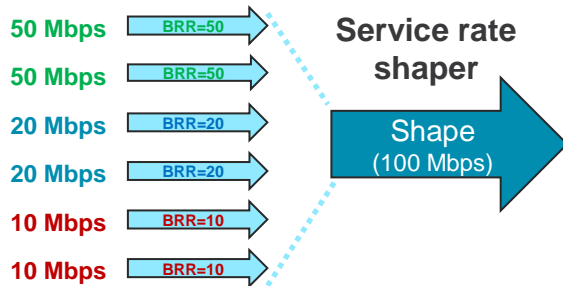


Bandwidth Remaining Ratio (BRR) provides proportional sharing to parent shapers during times of congestion.

If you over-subscribe your hub BR outbound bandwidth with per-tunnel policies that exceed the service rate, the BRR commands on each parent policy means they will get their “fair share” of the remaining bandwidth as compared to the other branch sites.

- If all the per-tunnel BW amounts are 5 Mbps or greater, we use a BRR value of $BW / 1 \text{ Mbps}$. (i.e. 10 Mbps is BRR of 10, 50 Mbps is BRR of 50, etc.)
- If any of the per-tunnel BW values are less than 5 Mbps, we use a BRR value of $BW / 100 \text{ Kbps}$. (i.e. 3 Mbps is BRR of 30, 1.5 Mbps is BRR of 15, etc.)

Per-Tunnel shapers



When the total bandwidth exceeds 100 Mbps, each of the per-tunnel shapers will get their fair share based on their BRR values.

Example:

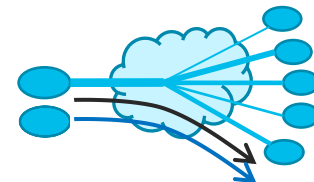
50 Mbps site gets $50 / 160$ or 31.25%

20 Mbps site gets $20 / 160$ or 12.5%

10 Mbps site gets $10 / 160$ or 6.25%

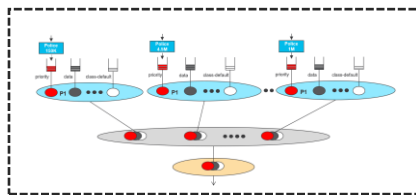
Multiple Sender QoS for Hub Routers

Bandwidth Sharing Between Multiple Senders



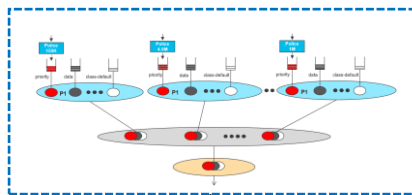
- Bandwidth can exceed 100% of the remote-site inbound Service Rate using a calculated oversubscription of ~ 1.6:1
- Bandwidth has to be divided equally due to one NHRP group
- QoS child policies do not have to be the same per Sender but DSCP markings must match for PfR TC channels to establish
- As the number of senders increase, the percentages need to come down accordingly based on the network administrators knowledge of their traffic patterns

Hub BR1 (MNH/MDC)



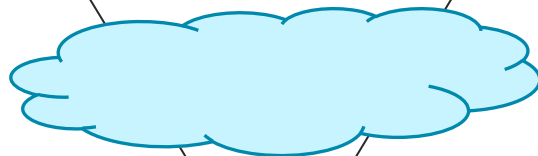
80% BW

Hub BR2 (MNH/MDC)



80% BW

- Total bandwidth should not exceed 160% of remote-site inbound Service Rate



To avoid unwanted SP drops of voice traffic, priority traffic from all senders should not exceed the remote site inbound service rate

Remote Site
Inbound Service Rate

Remote Site Example:

50 Mb/s * .80 = 40 Mb/s per Hub BR

Branch Tunnel Interface

```
interface Tunnel10
bandwidth receive 50000
nhrp nhs 10.6.34.1 nbma 192.168.6.1 multicast
nhrp nhs 10.6.34.2 nbma 192.168.6.41 multicast
nhrp group RS-GROUP-50MBPS-80
```

Hub Policy

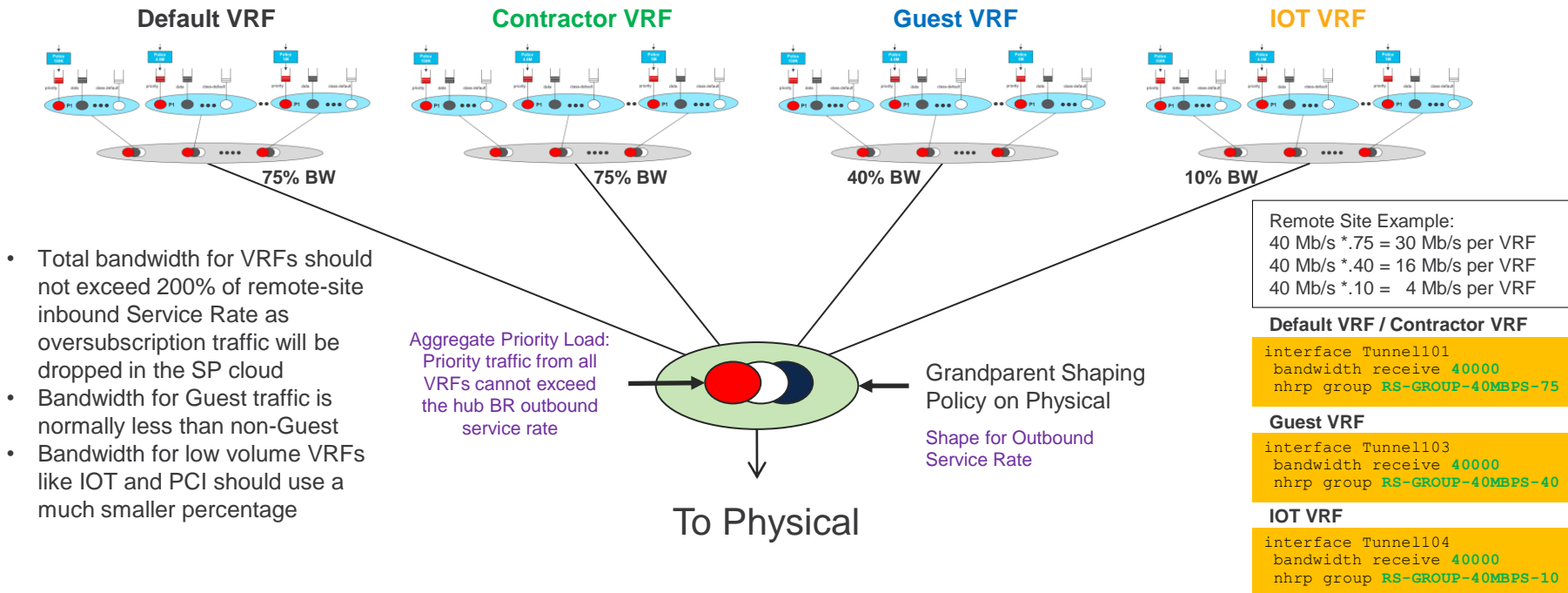
```
policy-map RS-GROUP-50MBPS-80-POLICY
class class-default
description 80% of 50 Mbps
shape average 40 Mbps
bandwidth remaining ratio 40
service-policy WAN
```

Multiple VRF QoS for Hub Routers

Bandwidth Sharing Between Multiple VRF Tunnels

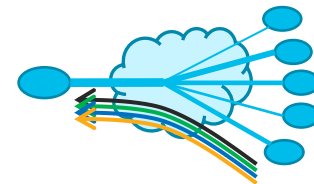


- Bandwidth can exceed 100% of the remote-site inbound Service Rate using an oversubscription ratio of ~ 2:1
- Bandwidth does not have to be divided equally between VRFs
- QoS policies do not have to be the same per VRF
- As the number of VRFs increase, the percentages need to come down accordingly based on the network administrators knowledge of their traffic patterns

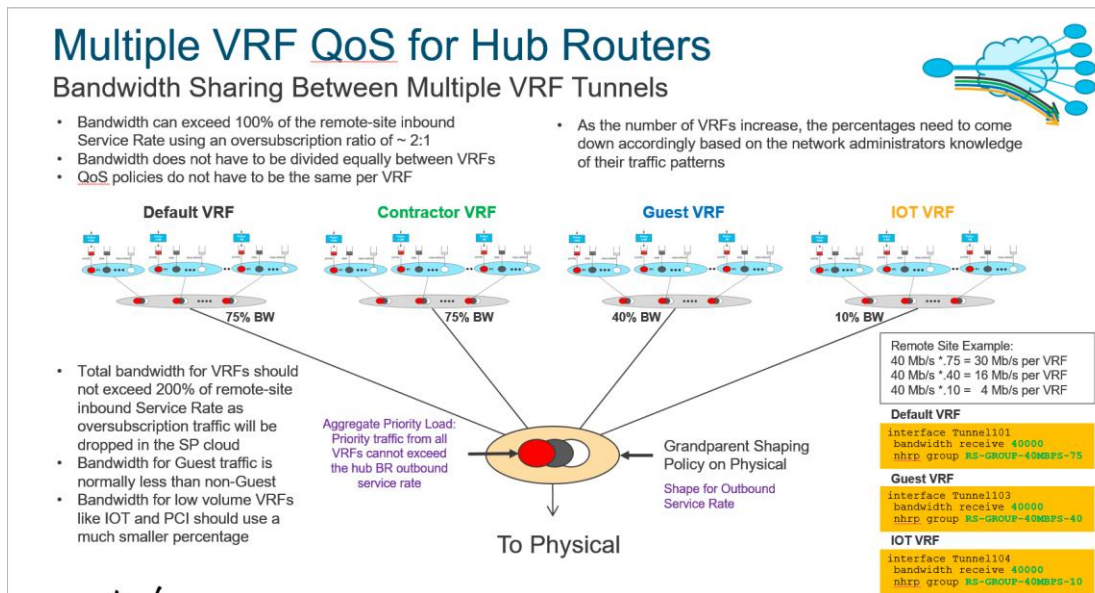


VRF = Virtual Routing and Forwarding

Multiple VRF QoS for Branch



- Using normal recommendations, QoS policy is applied to the physical interface at remote site which means all VRFs share the same QoS policy by default
- If you want to use different QoS policies for each VRF, you can deploy per-tunnel QoS in the spoke to hub direction using the same tools and limitations described on the previous slide



Enterprise to SP Mapping

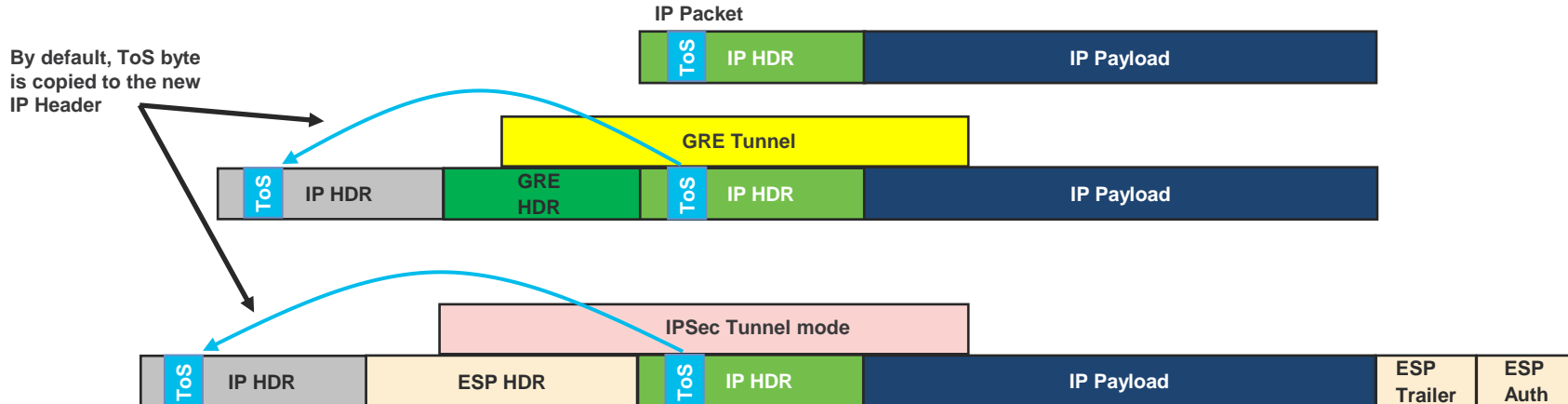
ToS Byte Preservation



The **12-class** view is preserved across the enterprise even though we treat it differently at the egress of the router and send it to different channels within the SP network

The **twelve classes remain intact** on the inner header and the outer tunnel header is remarked as the traffic leaves the tunnel interface

The remarked outer header is discarded after arriving at the tunnel interface on the receiving router, thus leaving the **inner header marking unchanged**



Enterprise to SP Mapping

Set dscp tunnel outbound on tunnel (Hub)

```
class-map match-all MULTIMEDIA_CONFERENCING-NBAR
match protocol attribute traffic-class multimedia-conferencing
match protocol attribute business-relevance business-relevant
```

```
policy-map INGRESS-MARKING
class MULTIMEDIA_CONFERENCING-NBAR
set dscp af41
```

Marking the
User IP header

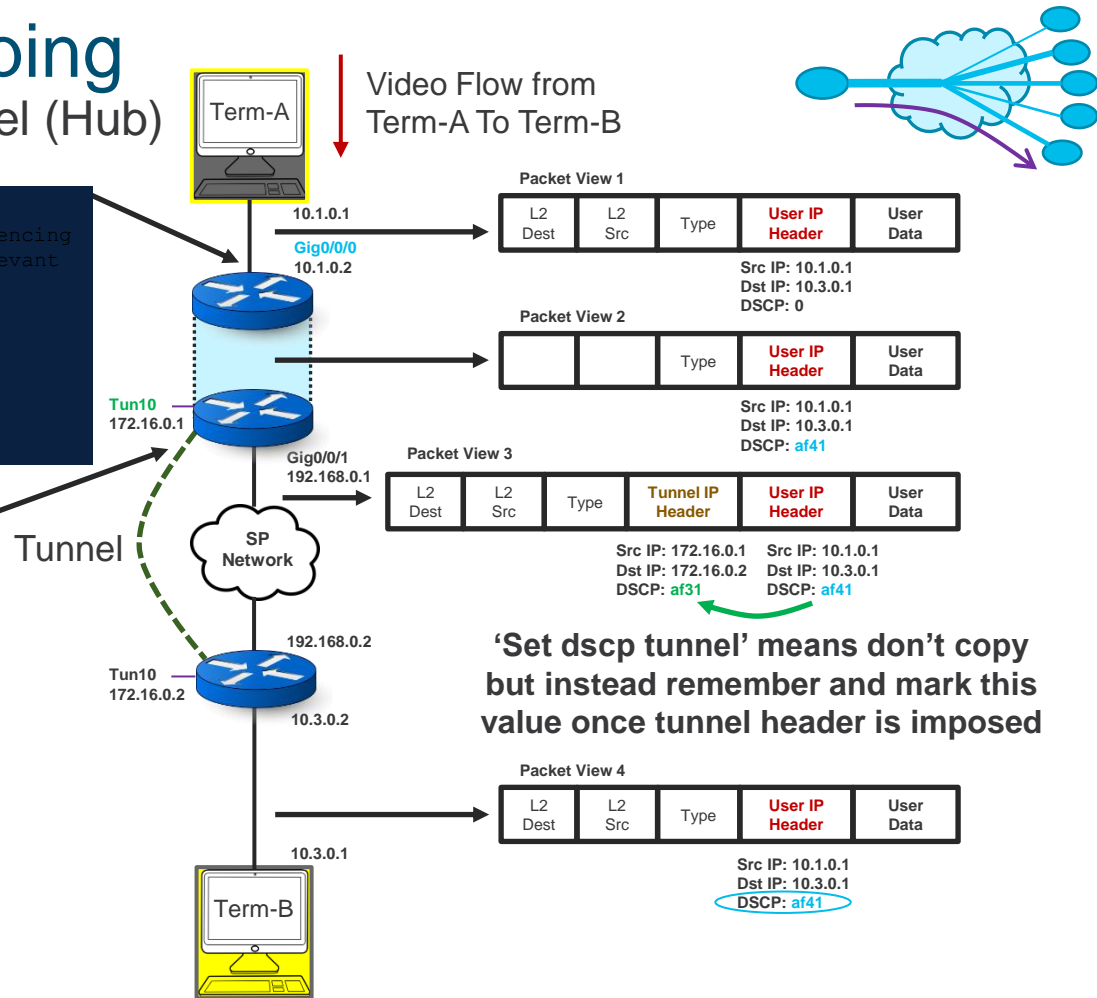
```
interface GigabitEthernet0/0/0
service-policy input INGRESS-MARKING
```

```
class-map INTERACTIVE-VIDEO
match dscp af41
```

```
policy-map RS-GROUP-10MBPS-POLICY
class INTERACTIVE-VIDEO
set dscp tunnel af31
```

Marking the
Tunnel IP header

```
interface Tunnel10
nhp map group RS-GROUP-10MBPS service-policy
output RS-GROUP-10MBPS-POLICY
```



Enterprise to SP Mapping

Set dscp outbound on physical (Branch)

```
class-map match-all MULTIMEDIA_CONFERENCING-NBAR
match protocol attribute traffic-class multimedia-conferencing
match protocol attribute business-relevance business-relevant
```

```
policy-map INGRESS-MARKING
class MULTIMEDIA_CONFERENCING-NBAR
set dscp af41
```

```
interface GigabitEthernet0/0/0
service-policy input INGRESS-MARKING
```

Marking the
User IP header

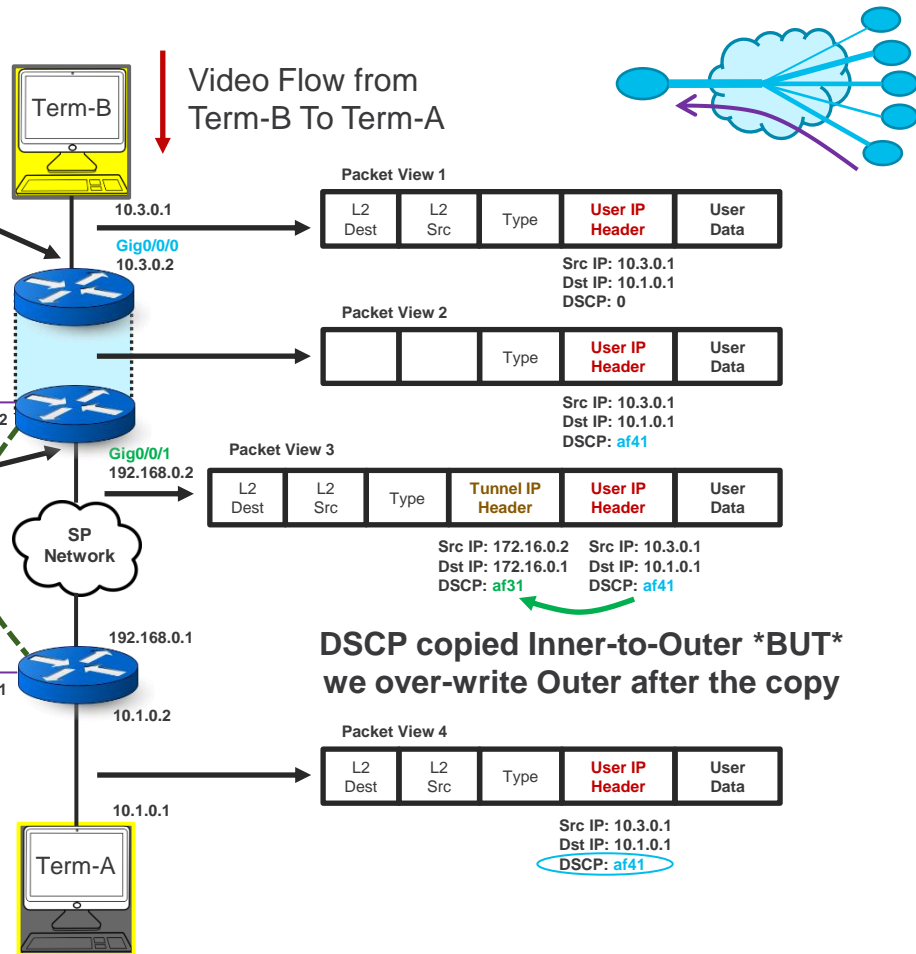
```
class-map INTERACTIVE-VIDEO
match dscp af41
```

```
policy-map POLICY-TRANSPORT-1
class INTERACTIVE-VIDEO
set dscp af31
```

```
interface GigabitEthernet0/0/1
service-policy output POLICY-TRANSPORT-1
```

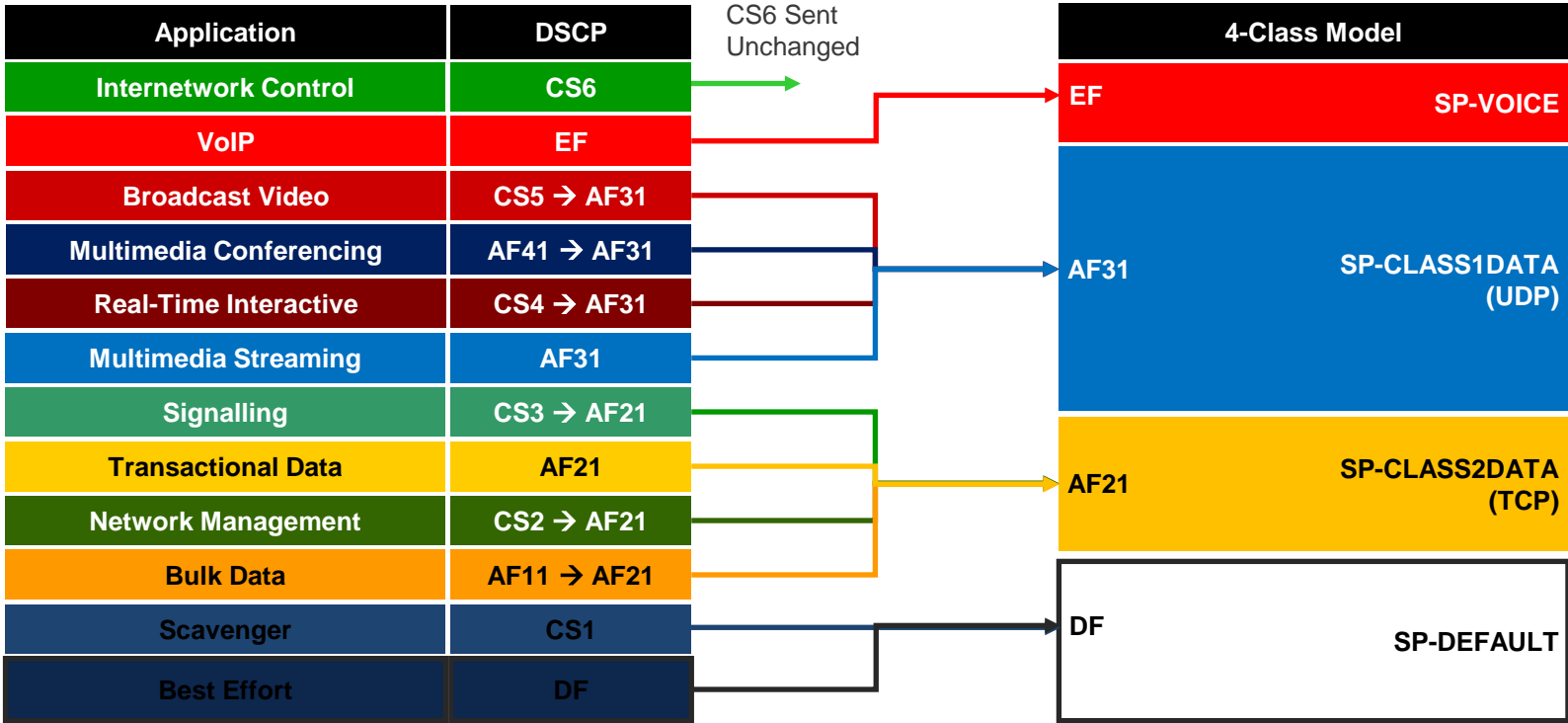
Marking the
Tunnel IP header

Tunnel



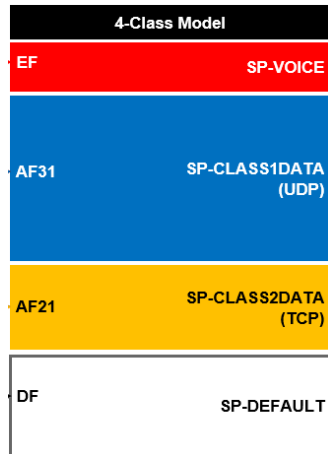
Enterprise to SP Mapping

Example: 4-Class SP Model



4-Class SP QoS Model Configuration

Tunnel Interface
Hub BR



```
policy-map WAN
class INTERACTIVE-VIDEO
bandwidth remaining percent 30
random-detect dscp-based
random-detect exponential-weighting-constant 9
set dscp tunnel af31
class STREAMING-VIDEO
bandwidth remaining percent 10
random-detect dscp-based
random-detect exponential-weighting-constant 9
set dscp tunnel af31
class NET-CTRL-MGMT
bandwidth remaining percent 5
set dscp tunnel cs6
class CALL-SIGNALING
bandwidth remaining percent 4
set dscp tunnel af21
class CRITICAL-DATA
bandwidth remaining percent 25
random-detect dscp-based
random-detect exponential-weighting-constant 9
set dscp tunnel af21
class SCAVENGER
bandwidth remaining percent 1
set dscp tunnel default
class VOICE
priority level 1
police cir percent 10
set dscp tunnel ef
class class-default
bandwidth remaining percent 25
random-detect
random-detect exponential-weighting-constant 9
set dscp tunnel default
```

Hub Router:

```
policy-map RS-GROUP-10MBPS-POLICY
class class-default
shape average 10 Mbps
bandwidth remaining ratio 10
service-policy WAN
```

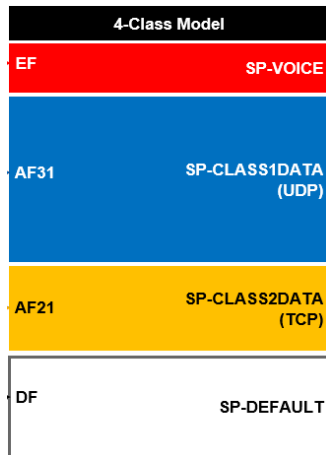
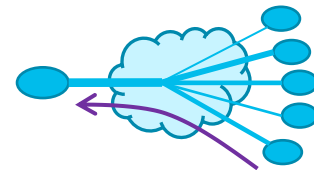
```
interface Tunnel10
bandwidth <service-rate>
nhp map group RS-GROUP-10MBPS service-policy
output RS-GROUP-10MBPS-POLICY
```

Branch Router:

```
interface GigabitEthernet0/0
bandwidth 10000
service-policy output POLICY-TRANSPORT-1
!
interface Tunnel10
bandwidth 10000
nhp group RS-GROUP-10MBPS
tunnel source GigabitEthernet0/0
tunnel vrf TRANSPORT-1
```

4-Class SP QoS Model Configuration

Physical Interface Branch



```
policy-map WAN
  class INTERACTIVE-VIDEO
    bandwidth remaining percent 30
    random-detect dscp-based
    random-detect exponential-weighting-constant 9
    set dscp af31
  class STREAMING-VIDEO
    bandwidth remaining percent 10
    random-detect dscp-based
    random-detect exponential-weighting-constant 9
    set dscp af31
  class NET-CTRL-MGMT
    bandwidth remaining percent 5
    set dscp cs6
  class CALL-SIGNALING
    bandwidth remaining percent 4
    set dscp af21
  class CRITICAL-DATA
    bandwidth remaining percent 25
    random-detect dscp-based
    random-detect exponential-weighting-constant 9
    set dscp af21
  class SCAVENGER
    bandwidth remaining percent 1
    set dscp default
  class VOICE
    priority level 1
    police cir percent 10
    set dscp ef
  class class-default
    bandwidth remaining percent 25
    random-detect
    random-detect exponential-weighting-constant 9
    set dscp default
```

Branch Router:

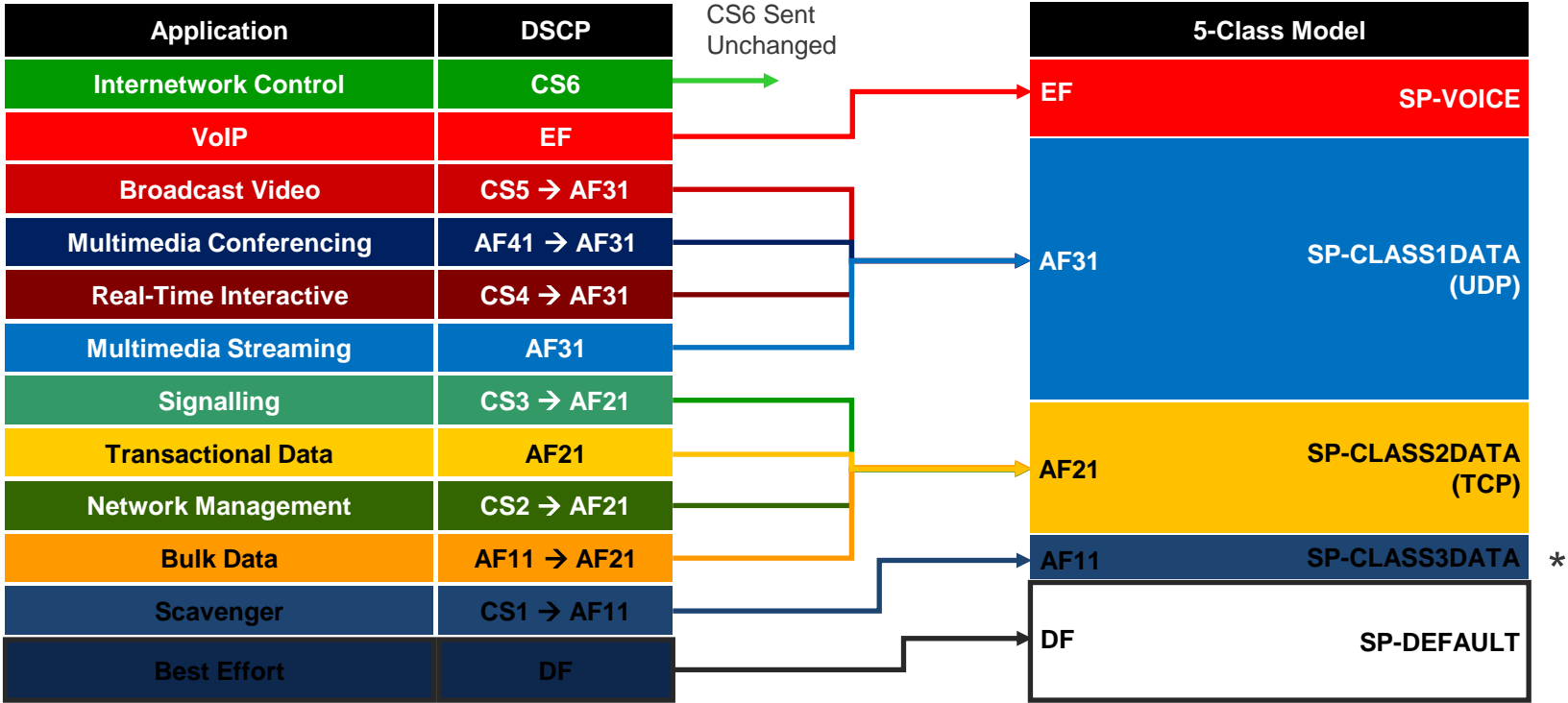
```
policy-map POLICY-TRANSPORT-1
  class class-default
    shape average 10 Mbps
    service-policy WAN
```

```
interface GigabitEthernet0/0
  bandwidth 10000
  service-policy output POLICY-TRANSPORT-1
```

The PfR Traffic Class channels will not establish if the DSCP values from the hub and branch routers do not match

Enterprise to SP Mapping

Example: 5-Class SP Model



* - Specified by ISP

5-Class QoS Model Configuration

Tunnel Interface Hub BR

Reference



5-Class Model	
EF	SP-VOICE
AF31	SP-CLASS1DATA (UDP)
AF21	SP-CLASS2DATA (TCP)
AF11	SP-CLASS3DATA
DF	SP-DEFAULT

```
policy-map WAN
class INTERACTIVE-VIDEO
bandwidth remaining percent 30
random-detect dscp-based
random-detect exponential-weighting-constant 9
set dscp tunnel af31
class STREAMING-VIDEO
bandwidth remaining percent 10
random-detect dscp-based
random-detect exponential-weighting-constant 9
set dscp tunnel af31
class NET-CTRL-MGMT
bandwidth remaining percent 5
set dscp tunnel cs6
class CALL-SIGNALING
bandwidth remaining percent 4
set dscp tunnel af21
class CRITICAL-DATA
bandwidth remaining percent 25
random-detect dscp-based
random-detect exponential-weighting-constant 9
set dscp tunnel af21
class SCAVENGER
bandwidth remaining percent 1
set dscp tunnel af11
class VOICE
priority level 1
police cir percent 10
set dscp tunnel ef
class class-default
bandwidth remaining percent 25
random-detect
random-detect exponential-weighting-constant 9
set dscp tunnel default
```

Hub Router:

```
policy-map RS-GROUP-10MBPS-POLICY
class class-default
shape average 10 Mbps
bandwidth remaining ratio 10
service-policy WAN
```

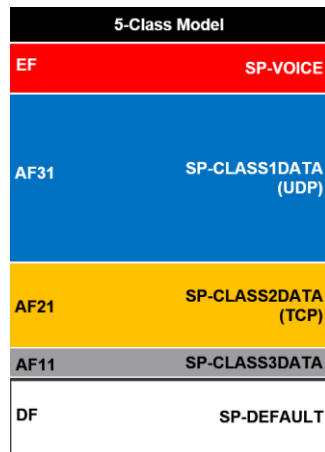
```
interface Tunnel10
bandwidth <service-rate>
nhp map group RS-GROUP-10MBPS service-policy
output RS-GROUP-10MBPS-POLICY
```

Branch Router:

```
interface GigabitEthernet0/0
bandwidth 10000
service-policy output POLICY-TRANSPORT-1
!
interface Tunnel10
bandwidth 10000
nhp group RS-GROUP-10MBPS
tunnel source GigabitEthernet0/0
tunnel vrf TRANSPORT-1
```

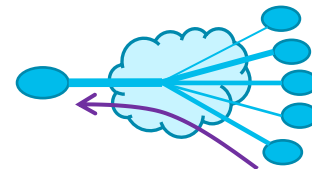
5-Class QoS Model Configuration

Physical Interface Branch



```
policy-map WAN
class INTERACTIVE-VIDEO
bandwidth remaining percent 30
random-detect dscp-based
random-detect exponential-weighting-constant 9
set dscp af31
class STREAMING-VIDEO
bandwidth remaining percent 10
random-detect dscp-based
random-detect exponential-weighting-constant 9
set dscp af31
class NET-CTRL-MGMT
bandwidth remaining percent 5
set dscp cs6
class CALL-SIGNALING
bandwidth remaining percent 4
set dscp af21
class CRITICAL-DATA
bandwidth remaining percent 25
random-detect dscp-based
random-detect exponential-weighting-constant 9
set dscp af21
class SCAVENGER
bandwidth remaining percent 1
set dscp af11
class VOICE
priority level 1
police cir percent 10
set dscp ef
class class-default
bandwidth remaining percent 25
random-detect
random-detect exponential-weighting-constant 9
set dscp default
```

Reference



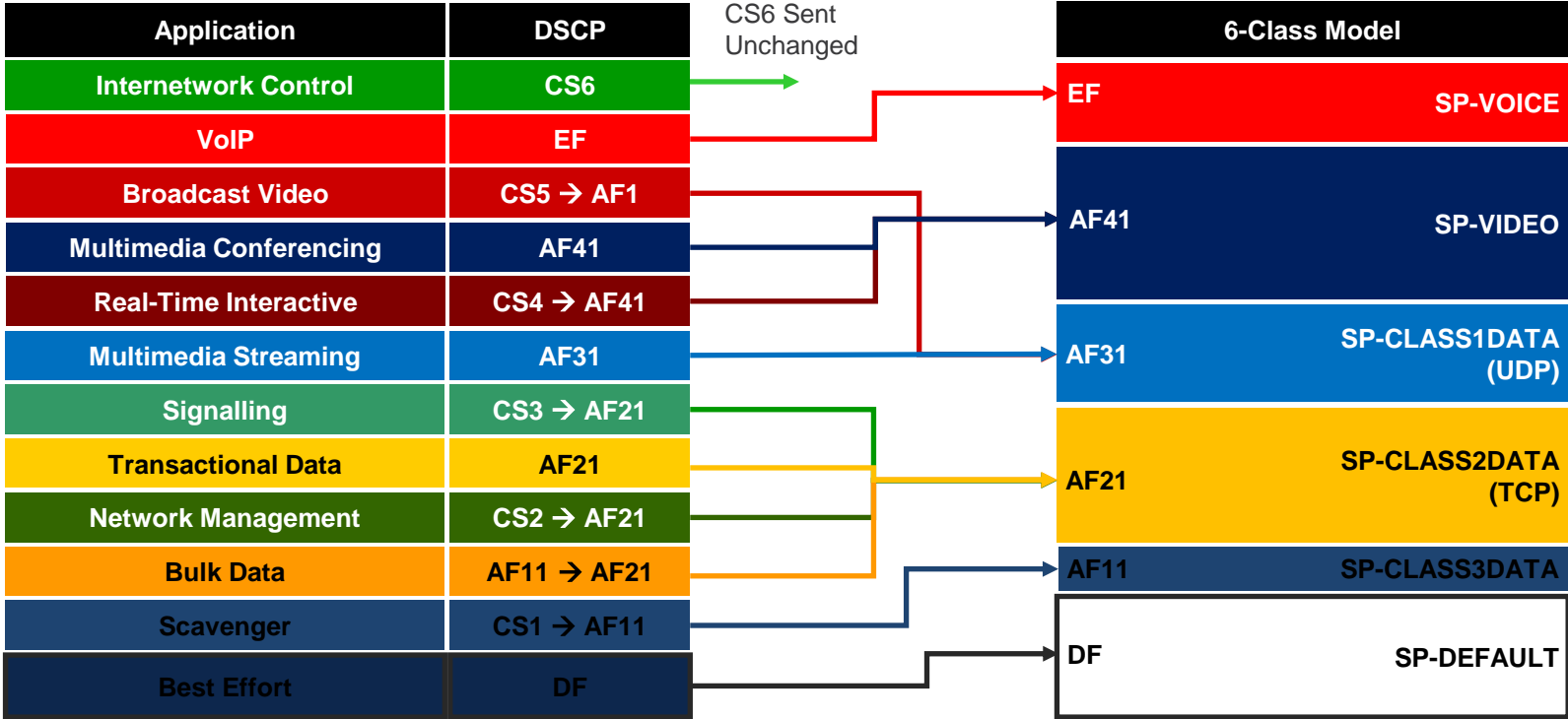
Branch Router:

```
policy-map POLICY-TRANSPORT-1
class class-default
shape average 10 Mbps
service-policy WAN
```

```
interface GigabitEthernet0/0
bandwidth 10000
service-policy output POLICY-TRANSPORT-1
```


Enterprise to SP Mapping

Example: 6-Class SP Model

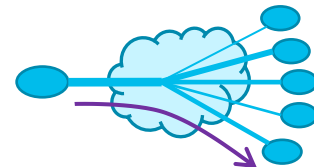


* - Specified by ISP

6-Class QoS Model Configuration

Tunnel Interface Hub BR

Reference



6-Class Model	
EF	SP-VOICE
AF41	SP-VIDEO
AF31	SP-CLASS1DATA (UDP)
AF21	SP-CLASS2DATA (TCP)
AF11	SP-CLASS3DATA
DF	SP-DEFAULT

```
policy-map WAN
class INTERACTIVE-VIDEO
bandwidth remaining percent 30
random-detect dscp-based
random-detect exponential-weighting-constant 9
set dscp tunnel af41
class STREAMING-VIDEO
bandwidth remaining percent 10
random-detect dscp-based
random-detect exponential-weighting-constant 9
set dscp tunnel af31
class NET-CTRL-MGMT
bandwidth remaining percent 5
set dscp tunnel cs6
class CALL-SIGNALING
bandwidth remaining percent 4
set dscp tunnel af21
class CRITICAL-DATA
bandwidth remaining percent 25
random-detect dscp-based
random-detect exponential-weighting-constant 9
set dscp tunnel af21
class SCAVENGER
bandwidth remaining percent 1
set dscp tunnel af11
class VOICE
priority level 1
police cir percent 10
set dscp tunnel ef
class class-default
bandwidth remaining percent 25
random-detect
random-detect exponential-weighting-constant 9
set dscp tunnel default
```

Hub Router:

```
policy-map RS-GROUP-10MBPS-POLICY
class class-default
shape average 10 Mbps
bandwidth remaining ratio 10
service-policy WAN
```

```
interface Tunnel10
bandwidth <service-rate>
nhp map group RS-GROUP-10MBPS service-policy
output RS-GROUP-10MBPS-POLICY
```

Branch Router:

```
interface GigabitEthernet0/0
bandwidth 10000
service-policy output POLICY-TRANSPORT-1
!
interface Tunnel10
bandwidth 10000
nhp group RS-GROUP-10MBPS
tunnel source GigabitEthernet0/0
tunnel vrf TRANSPORT-1
```

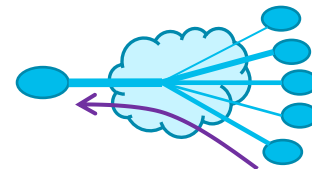
6-Class QoS Model Configuration

Physical Interface Branch

6-Class Model	
EF	SP-VOICE
AF41	SP-VIDEO
AF31	SP-CLASS1DATA (UDP)
AF21	SP-CLASS2DATA (TCP)
AF11	SP-CLASS3DATA
DF	SP-DEFAULT

```
policy-map WAN
class INTERACTIVE-VIDEO
bandwidth remaining percent 30
random-detect dscp-based
random-detect exponential-weighting-constant 9
set dscp af41
class STREAMING-VIDEO
bandwidth remaining percent 10
random-detect dscp-based
random-detect exponential-weighting-constant 9
set dscp af31
class NET-CTRL-MGMT
bandwidth remaining percent 5
set dscp cs6
class CALL-SIGNALING
bandwidth remaining percent 4
set dscp af21
class CRITICAL-DATA
bandwidth remaining percent 25
random-detect dscp-based
random-detect exponential-weighting-constant 9
set dscp af21
class SCAVENGER
bandwidth remaining percent 1
set dscp af11
class VOICE
priority level 1
police cir percent 10
set dscp ef
class class-default
bandwidth remaining percent 25
random-detect
random-detect exponential-weighting-constant 9
set dscp default
```

Reference



Branch Router:

```
policy-map POLICY-TRANSPORT-1
class class-default
shape average 10 Mbps
service-policy WAN
```

```
interface GigabitEthernet0/0
bandwidth 10000
service-policy output POLICY-TRANSPORT-1
```