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Webex Bandwidth Management

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BRKCOL-2777



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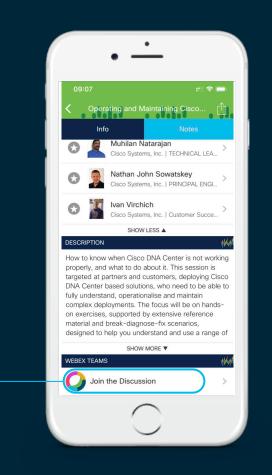
Cisco Webex Teams

Questions?

Use Cisco Webex Teams to chat with the speaker after the session

How

- 1 Find this session in the Cisco Events Mobile App
- 2 Click "Join the Discussion" -
- 3 Install Webex Teams or go directly to the team space
- 4 Enter messages/questions in the team space



Agenda

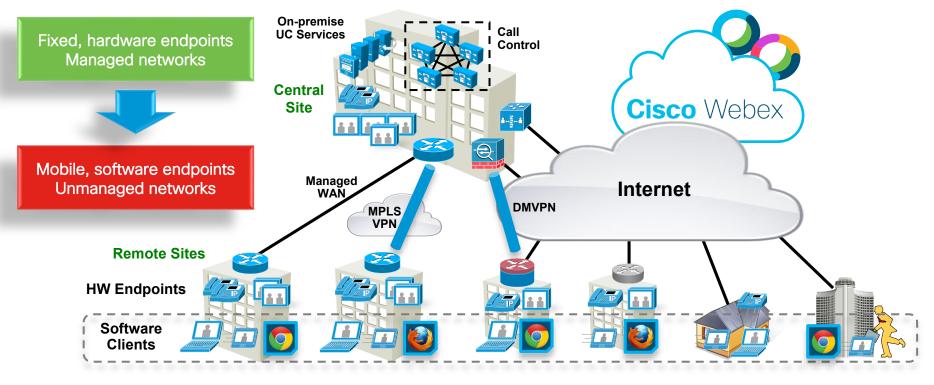
- Introduction
- QoS Strategy Overview
- QoS Marking for Webex Media
- Bandwidth Provisioning for Webex Media

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Introduction

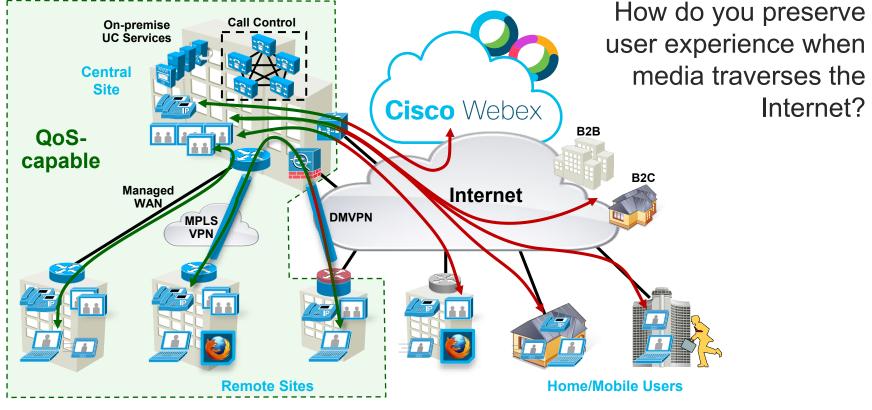


Evolution of Collaboration Landscape

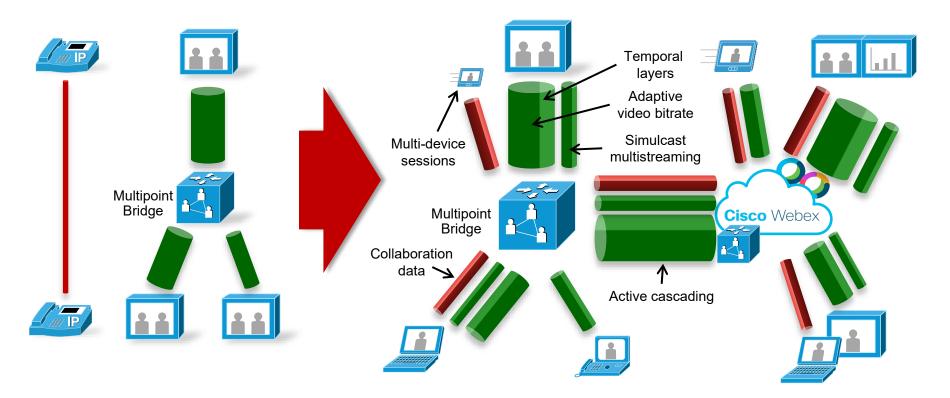


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Managed vs. Unmanaged Networks Where do your media packets go?



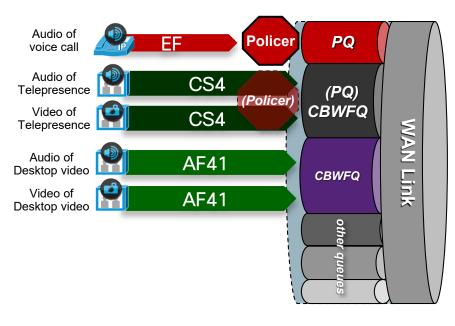
Evolution of Collaboration Media Streams



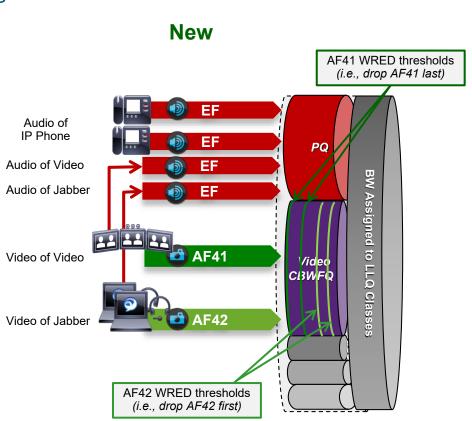
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QoS Tools Evolution of Queuing Recommendations

Previous

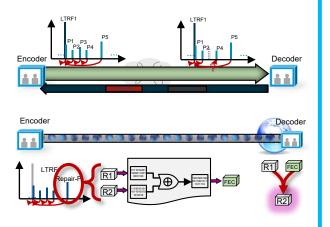






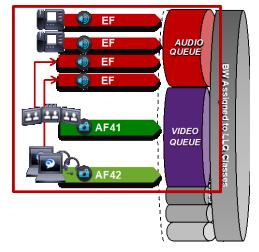
Our Strategy

"Smart" Media Techniques



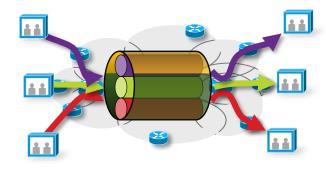
- Use media resilience to reduce impact of packet loss
- Apply rate adaptation to reduce network congestion

Commonly Deployed QoS Tools



- Consolidate mechanisms to identify Collaboration media
- Evolve classification and scheduling recommendations

New Design & Deployment Guidelines



Leverage media resilience and rate adaptation to enable pervasive video deployments through:

- simplified provisioning
- optimized bandwidth utilization

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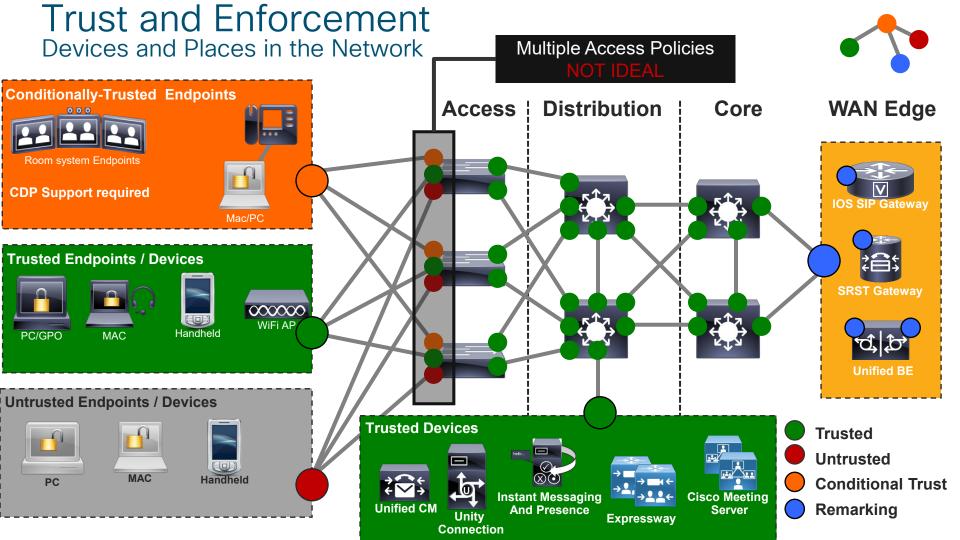


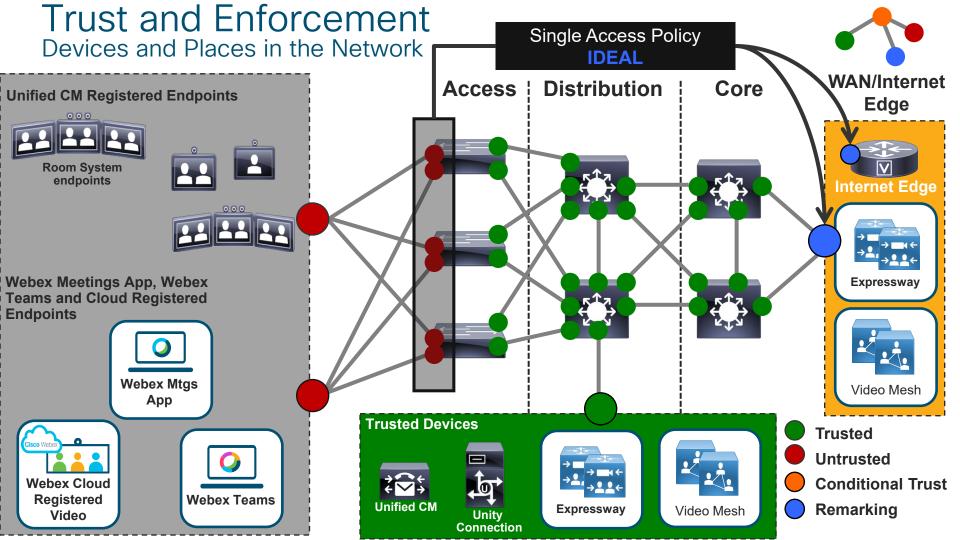
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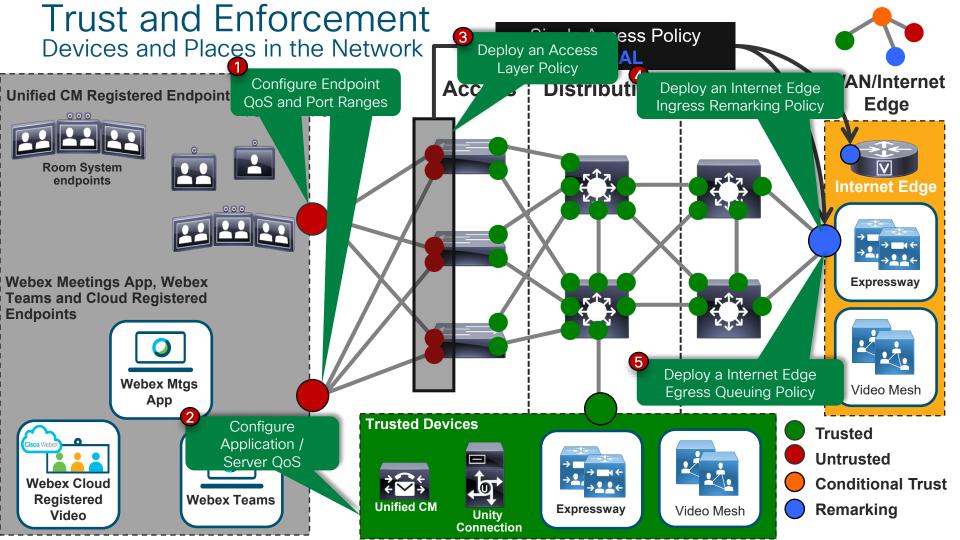
Classification: DSCP Classes

- EF: Expedited Forwarding (PQ)
 - Used for voice media
- AF: Assured Forwarding (CWBFQ)
 - Used for video media
- CS: Class Selector
 - Used for signaling

DSCP Class	DSCP	ToS Prec.
none	0	0
CS1	8	1
AF11	10	1
AF12	12	1
AF13	14	1
CS2	16	2
AF21	18	2
AF22	20	2
AF23	22	2
CS3	24	3
AF31	26	3
AF32	28	3
AF33	30	3
CS4	32	4
AF41	34	. 4
AF42	36	4
AF43	38	4
CS5	40	5
EF	46	5
CS6	48	6
CS7	56	7







Dual video queue

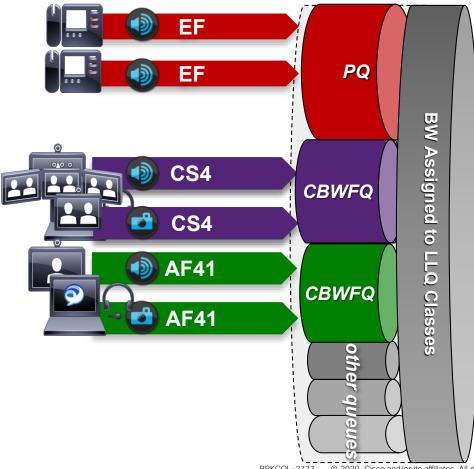
Audio of IP Phone

Audio of TelePresence

Video of TelePresence

Audio of Desktop Video

Video of Desktop Video



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Single video queue

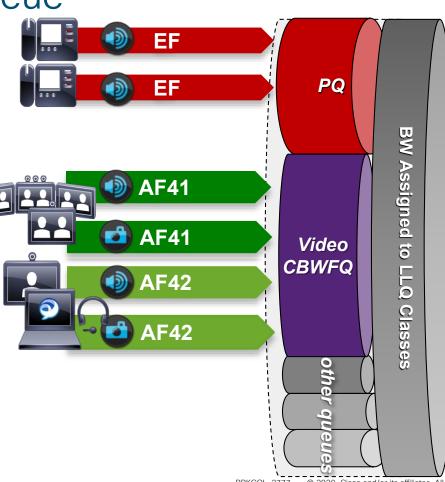
Audio of IP Phone

Audio of TelePresence

Video of TelePresence

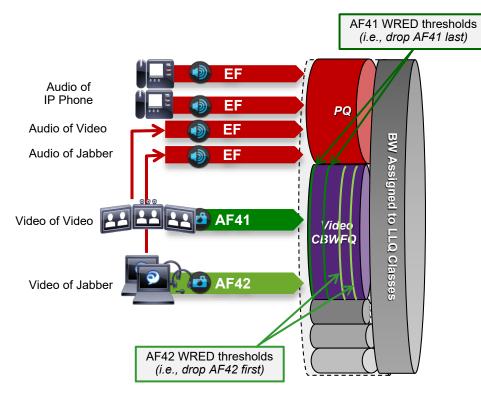
Audio of Desktop Video

Video of Desktop Video



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WAN Queuing Considerations Summary



- Map audio streams of voice and video calls (EF) to a priority queue
- Map video streams of video calls (AF41 and AF42) to a classbased queue with WRED:
 - AF41: higher drop thresholds (e.g., 50-100% of queue depth)
 - AF42: lower drop thresholds (e.g., 15-35% of queue depth)
- During congestion, AF42 traffic is dropped first:
 - Packet loss triggers rate adaptation
 - Media resilience limits the impact

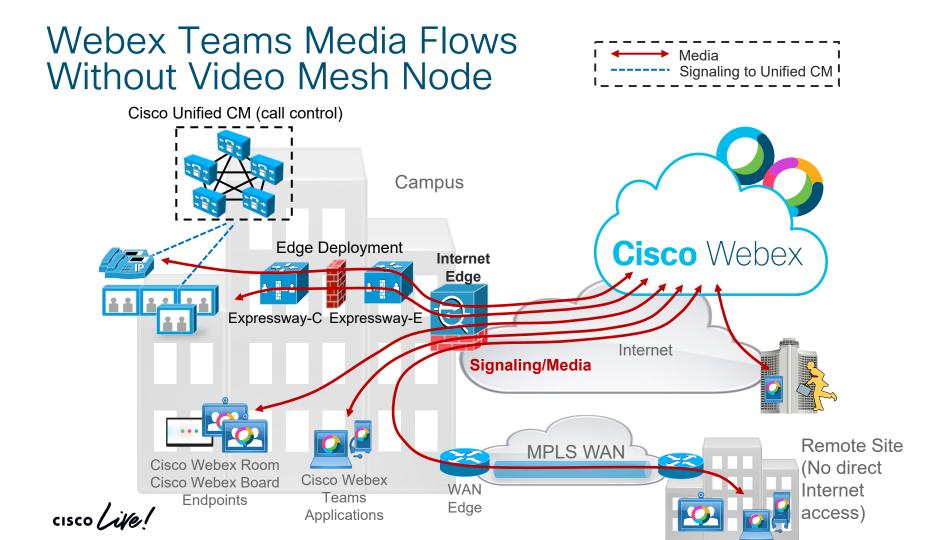


- Create a Simplified QoS Policy Use throughout the network
- Egress Queuing Policy
 - Single video queue for AF class traffic model is recommended
 - Consider multiple classes of Video AF41, AF42, AF43 (Prioritized Video vs Opportunistic Video)



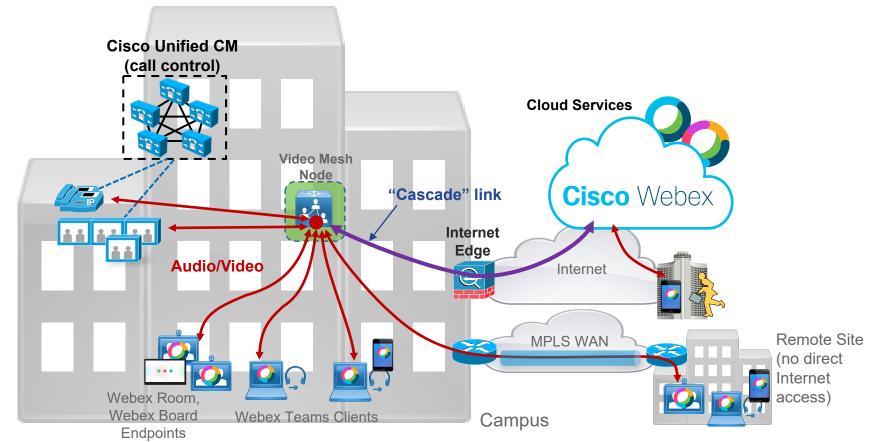






Webex Teams Media Flows With Video Mesh Node

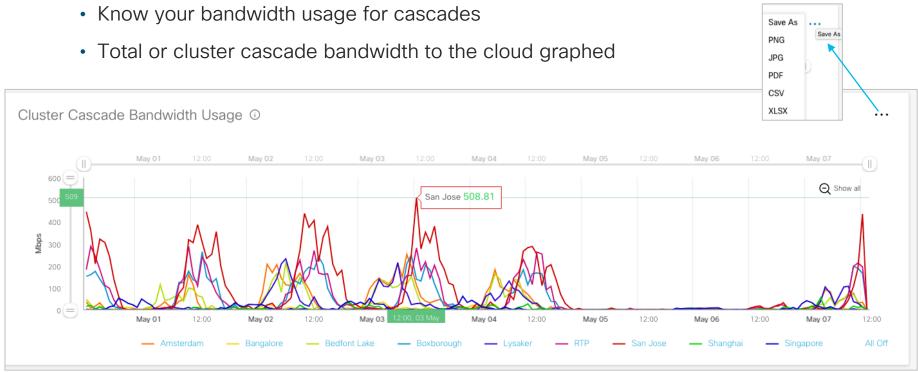




Cascade Bandwidth

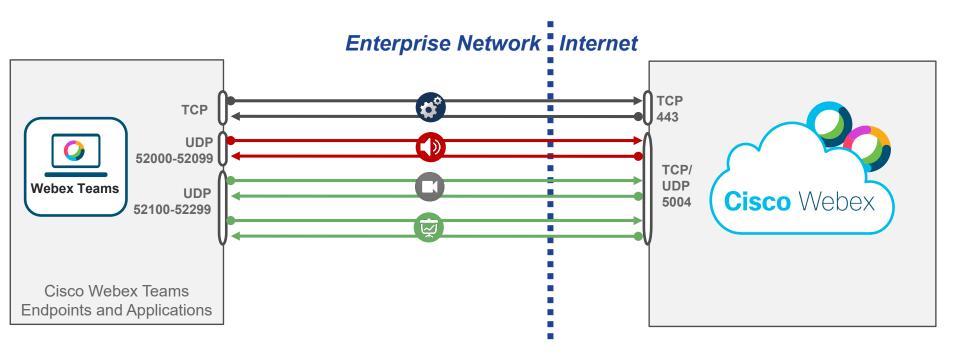
- Cascade bandwidth graphs are available in Control Hub.
 - The graphs are a per-cluster measurement.
- Cascade bandwidth per meeting is not configurable by the customer in Cisco Webex Control Hub.
- Maximum per-meeting negotiated cascade bandwidth is 20 Mbps for main video for all sources and the multiple main video streams they could send. This does not include the content channel or audio.
- In a 3 month timeframe with the top 20 customers based on the number of meetings with an average of 9100 meetings and 15,000 calls per month, the average total cascade bandwidth (Tx + Rx) for all the meetings was 11.6 Mbps.
- Webex Meetings new 5x5 grid layout is now supported on the Video Mesh.

Cascade Bandwidth Reports



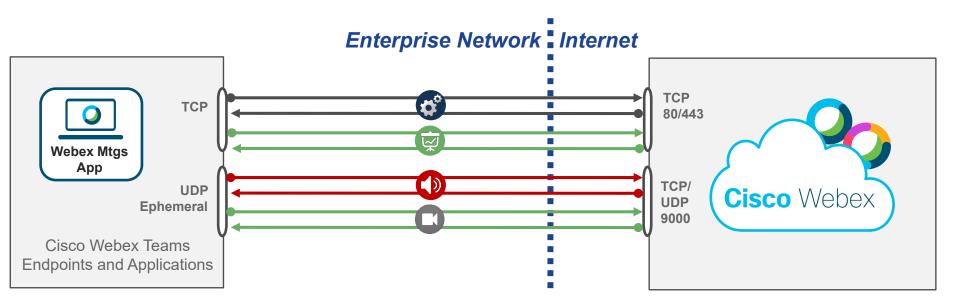
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Port Usage Today – Webex Teams



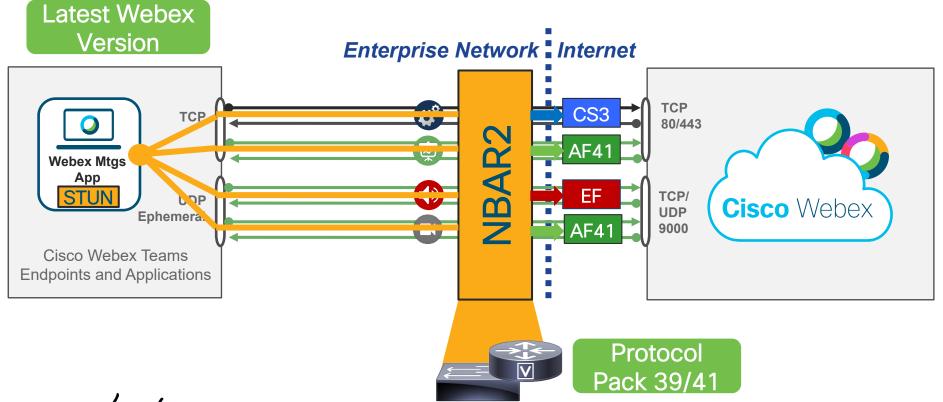
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Port Usage Today – Webex Meetings App



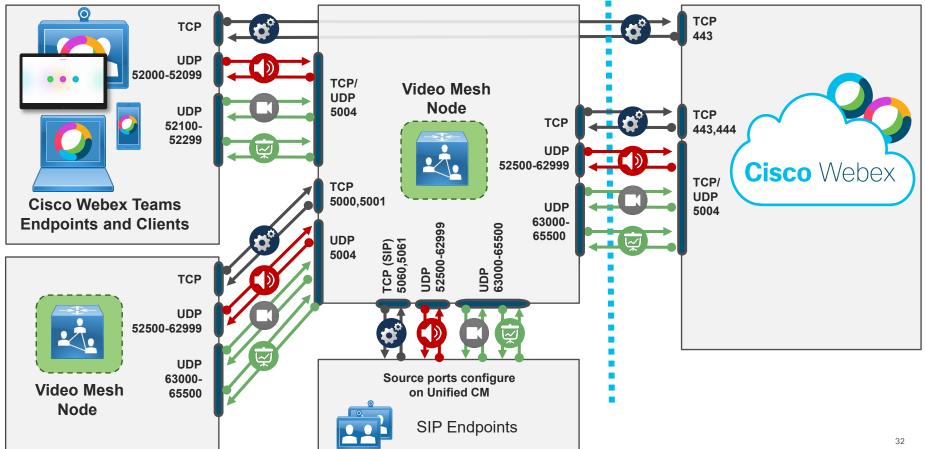
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Port Usage Today – Webex Meetings App



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Port Usage Today – Webex Teams + Video Mesh Enterprise Network: Internet



Expressway

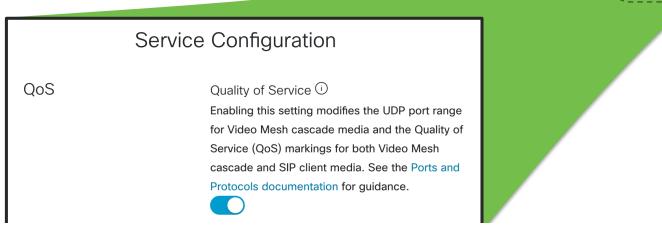
- System \rightarrow Quality of Service
 - DSCP Signaling value 24 (Default) → CS3
 - DSCP Audio value 46 (Default) → EF
 - DSCP Video value 34 (Default) → AF41
 - DSCP XMPP value 24 (Default) → CS3

	Status System	Configuration Applications	Users	Maintenance	<u>▲</u> ? œ
(Quality of Service			You are here: System	Quality of Service
	Tagging				
	DSCP Signaling value	* 24 👔			
	DSCP Audio value	* 46 (i)			
	DSCP Video value	* 34 (i)			
	DSCP XMPP value	* 24 👔			

Cisco Expressway Core (C) and Edge (E)

Video Mesh Node - Ports

- Webex Control Hub Services > Video Mesh
- QoS (Enabled by Default)
- Enables Cascade Port Ranges and Native Marking
 - Audio 52500-62999 (EF)*
 - Video 63000-65500 (AF41)*



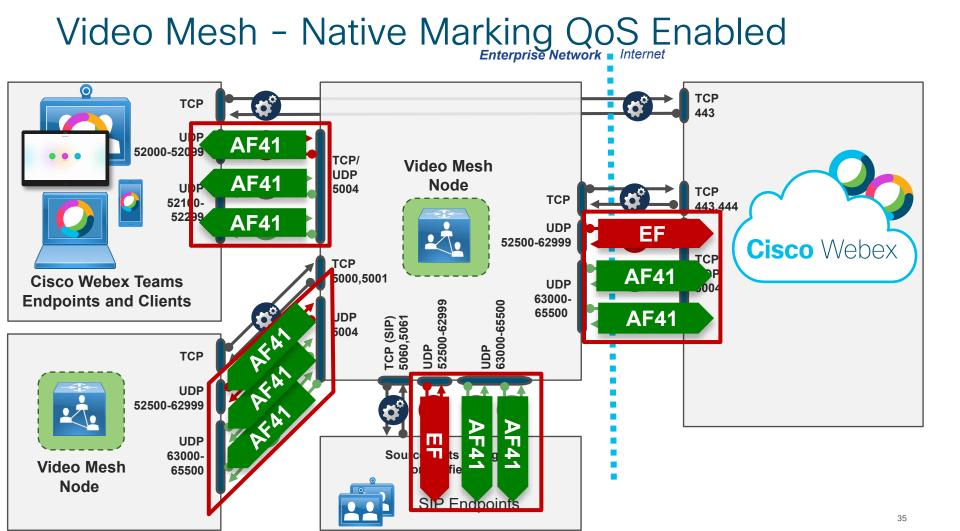
Pererence

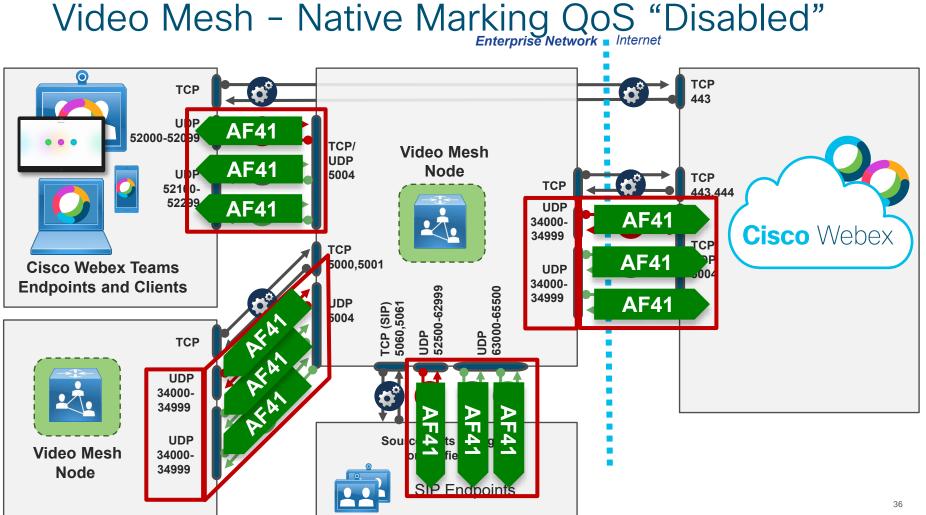
Video Mesh Cluster

Video Mesh

Node

* When disabled changes the source ports that are used for audio, video, and content sharing from the Video Mesh node to the range 34000 to 34999





Media Signatures for Cisco Webex Teams and Cloud Registered Endpoints Client to Cloud (Reverse for Cloud to Client)



Source IP	Destination IP	Source UDP Ports	Destination UDP Ports	Recommended DSCP	Media Type ³
Webex Teams application or endpoint	Webex cloud and Video Mesh Media Services	52000 to 52099	5004	EF	Audio
Webex Teams application or endpoint	Webex cloud and Video Mesh Media Services	52100 to 52299	5004	AF41	Video
Video Mesh Node	Webex Cloud Media Services	52500 to 62999	5004	EF	Audio
Video Mesh Node	Webex cloud Media Services	63000 to 65500	5004	AF41	Video
Video Mesh Node	Video Mesh Node	52500 to 62999	5004	EF	Audio
Video Mesh Node	Video Mesh Node	63000 to 65500	5004	AF41	Video

Media Signatures for Cisco Webex Meetings Application Client to Cloud (Reverse for Cloud to Client)

Source IP	Destination IP	Source UDP Ports	Destination UDP Ports	Recommended DSCP	Media Type
Cisco Webex Meetings Application	Webex Cloud	Ephemeral	9000	AF41	Audio / Video

Webex Meetings App

webex-audio = Webex audio streaming webex-video = Webex video streaming webex-app-sharing = Webex app sharing traffic webex-meeting = Webex Signaling Traffic - (nonmedia – not port based)



PERFERENCE



ip access-list extended QOS_VOICE
 permit udp any range 17000 17999 any dscp ef
ip access-list extended QOS_PRIORITIZED_VIDEO
 permit udp any range 17000 17999 any dscp af41
ip access-list extended QOS_WEBEX_TEAMS_AUDIO
 permit udp any range 52000 52099 any any
 permit udp any eq 5004 any range 52000 52099
ip access-list extended QOS_WEBEX_TEAMS_VIDEO
 permit udp any range 52100 52299 any any
 permit udp any eq 5004 any range 52100 52299

class-map match-any **VOICE**

match pccessedrWEBEKameDQOS_VOICE

match access-group name QOS_WEBEX_TEAMS_AUDIO

class-map match-any **PRIORITIZED_VIDEO**

match processedrwepEKaMeDEOS_PRIORITIZED_VIDEO match processedrwepEKamePOOSAWEBEX_TEAMS_VIDEO

Webex Ingress QoS Policy



ip access-list extended QOS_VOICE
 permit udp any range 17000 17999 any dscp ef
ip access-list extended QOS_PRIORITIZED_VIDEO
 permit udp any range 17000 17999 any dscp af41
ip access-list extended QOS_WEBEX_TEAMS_AUDIO
 permit udp any range 52000 52099 any any
 permit udp any eq 5004 any range 52000 52099
ip access-list extended QOS_WEBEX_TEAMS_VIDEO
 permit udp any range 52100 52299 any any
 permit udp any eq 5004 any range 52100 52299

class-map match-any **VOICE**

match protessebroupcoaspa00SaV016E

match access-group name QOS_WEBEX_TEAMS_AUDIO

class-map match-any **PRIORITIZED_VIDEO**

match accessedroupcnameagosviddeoRITIZED_VIDEO

match access-group name QOS WEBEX TEAMS VIDEO

WAN Ingress QoS Marking Policy

Ingress Policy '

Pererence



! This section applies the policy-map to the Interface Router(config-if)# service-policy input INGRESS-MARKING ! Attaches service policy to interface

3

! This section configures the classes class-map match-any VOICE match webex-audio match cisco-spark-audio class-map match-any PRIORITIZED-VIDEO match webex-video class-map match-any SIGNALING match webex-meeting match cisco-spark ! This section configures the policy-map to set DSCP for Trusted and Untrusted Voice, Video and SIP Signaling on ingress

policy-maxINGRESS-MARKING

clas

set dscp ef

class PRIORITIZED-VIDEO

set dscp af41

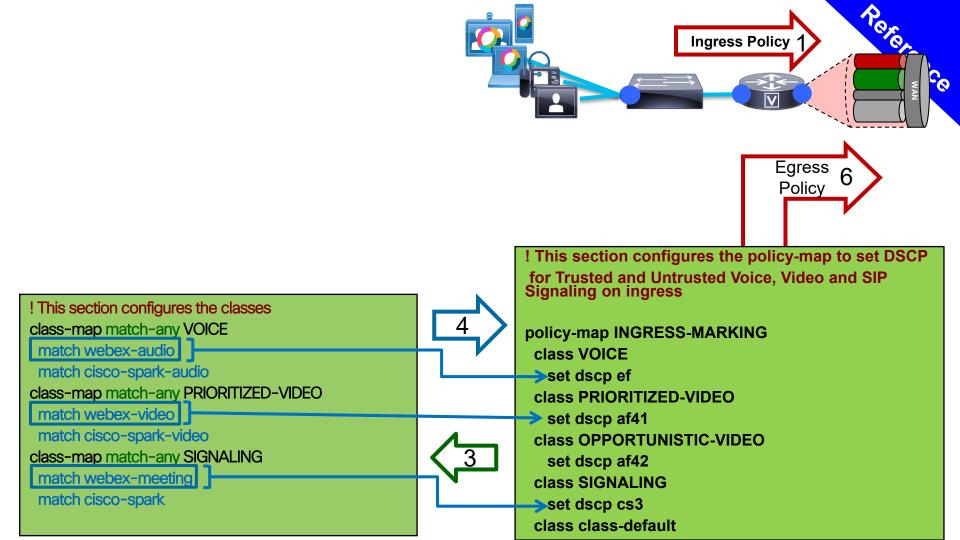
class OPPORTUNISTIC-VIDEO

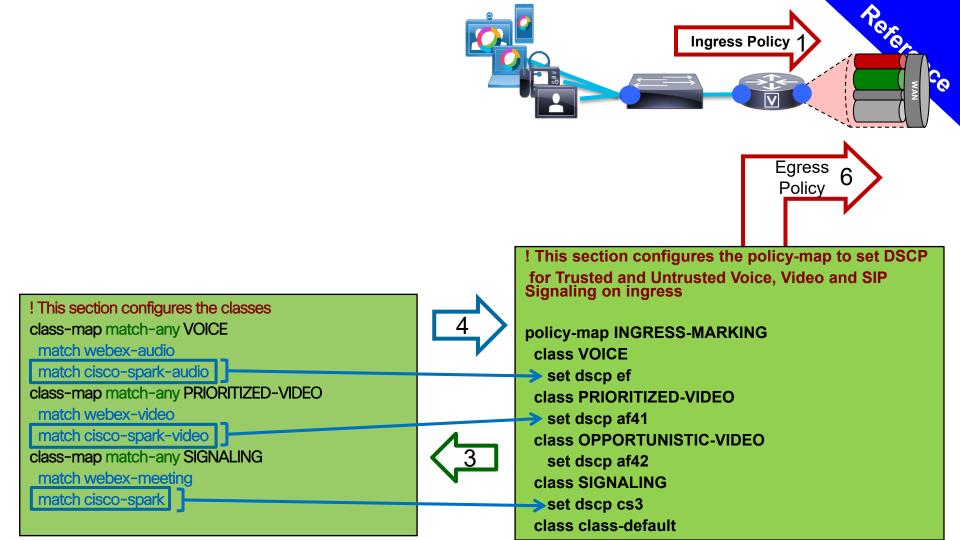
set dscp af42

class SIGNALING

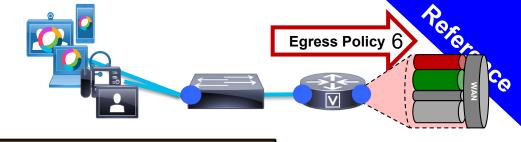
set dscp cs3

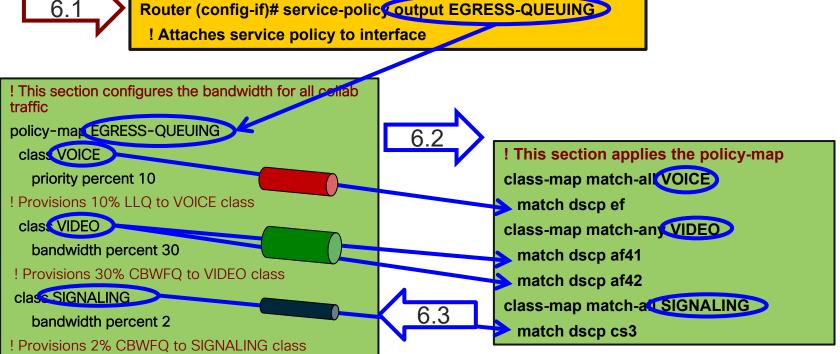
class class-default



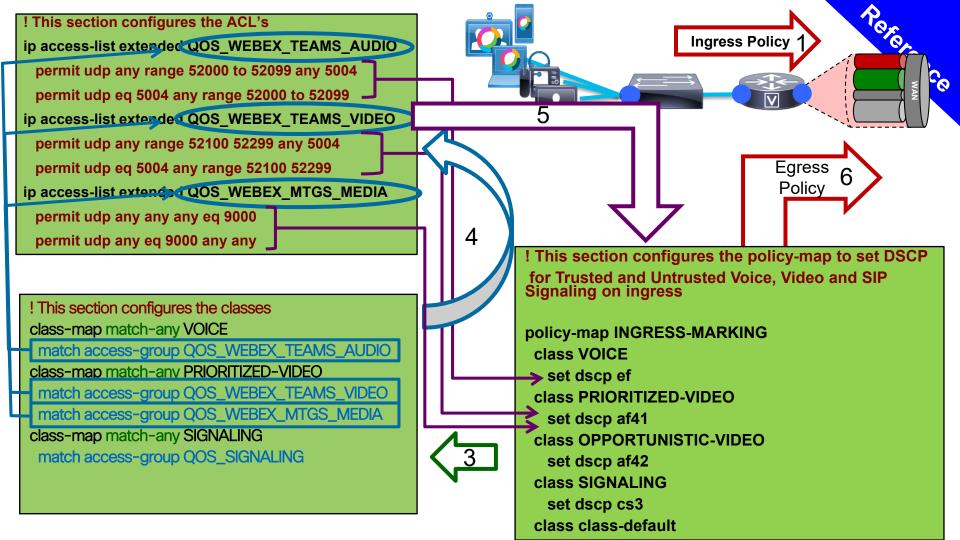


Egress Classification and Queuing



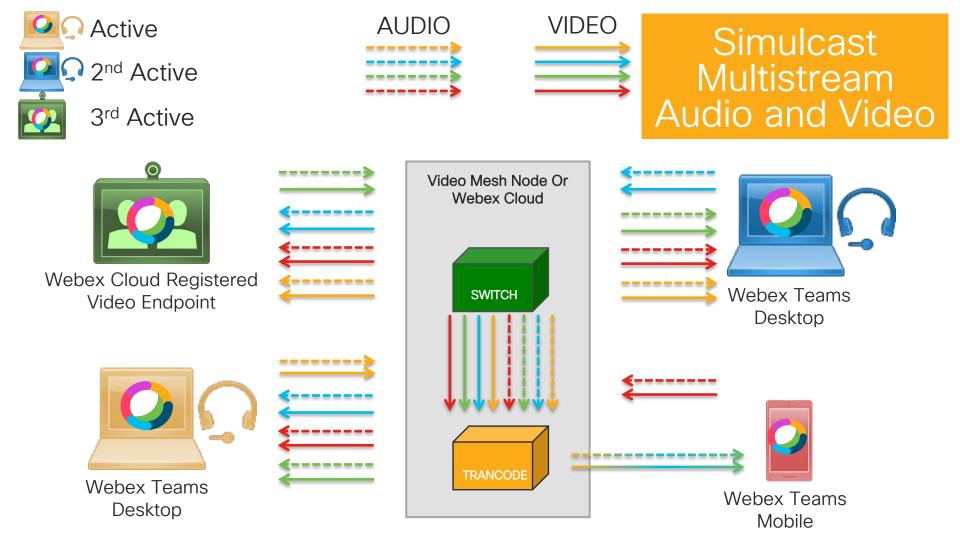


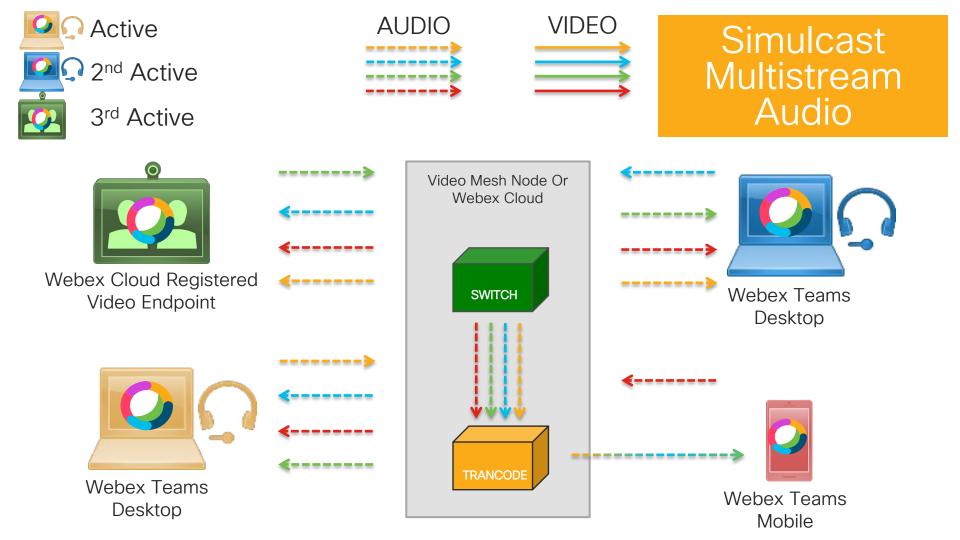
This section applies the policy-map to the Interface



Bandwidth Provisioning for Webex Media







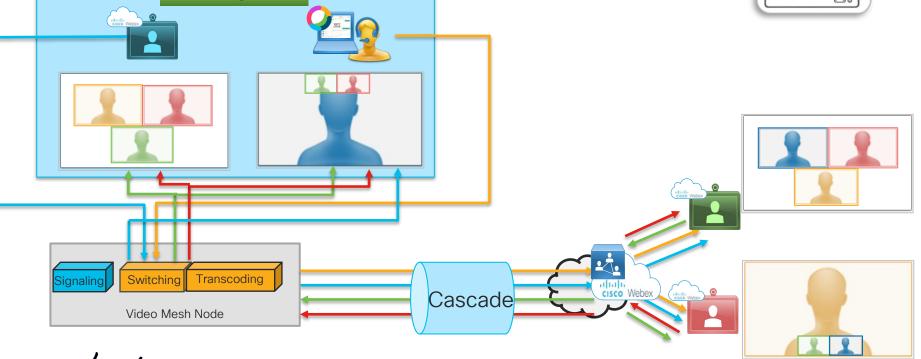
Main Video with Multiple Layouts Example

Cloud Registered



Active Presence and Equal View Layout





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Webex Multistream Audio and Video Summary

- Audio uses the Opus codec which allows for rate adaptation
- Audio of the last 3 active speakers sent to Webex Teams and Webex Meetings App Desktop Clients as well as Cloud Registered Endpoints.
- Webex Teams and Meetings Mobile Apps are transcoded limiting audio and video streaming. Regular audio video streaming, not multistream capable.

Caveats and Expectations

- Bandwidth Provisioning Variability
 - Multistream Video: Multiple send and receive resolutions based on requested layouts!
 - Multistream Audio: Affects receive rate. Up to 3 audio streams!
 - Media Assure: FEC, Rate Adaptation, LTRF, GDR, Packet Pacing...
- The greater the input, the larger the buffer of spikes, the more bandwidth available for calls that exceed expectations.
 - Otherwise said: As the number of users/concurrent calls goes up, the more the bandwidth utilization will tend smooth out!
- You'll see various numbers in documentation and it doesn't mean they are wrong! The values here are simplified as a starting point!

Approximate Bandwidth Values 720p For Provisioning Purposes – Avg values

- UCM Registered SIP endpoints Expressway sizing
 - Voice call (or audio stream of a video call): 80 kbps
 - Video streams of video endpoints: 2mb at 720p resolution
 - Use Video Mesh sizing for UCM registered endpoints

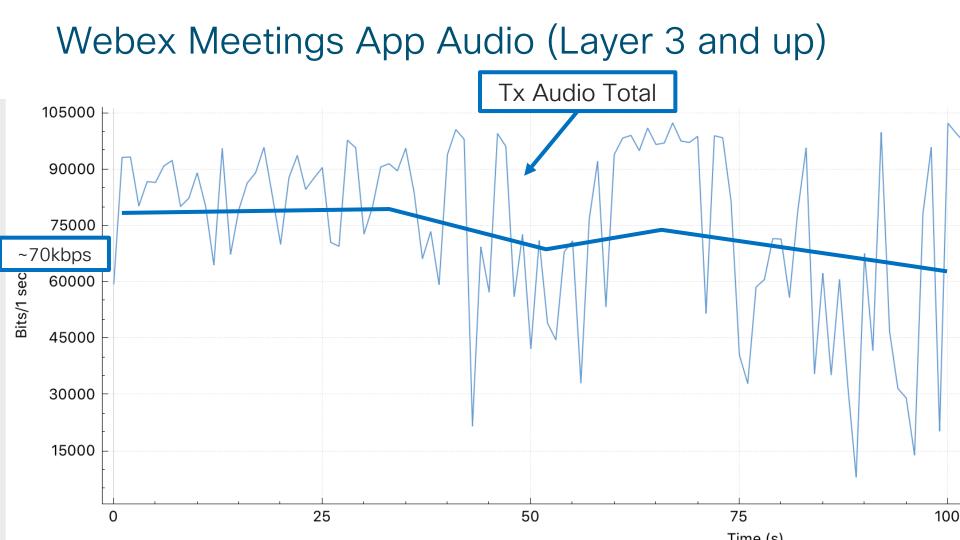
Approximate Video Bandwidth Values 720p For Provisioning Purposes – Avg values

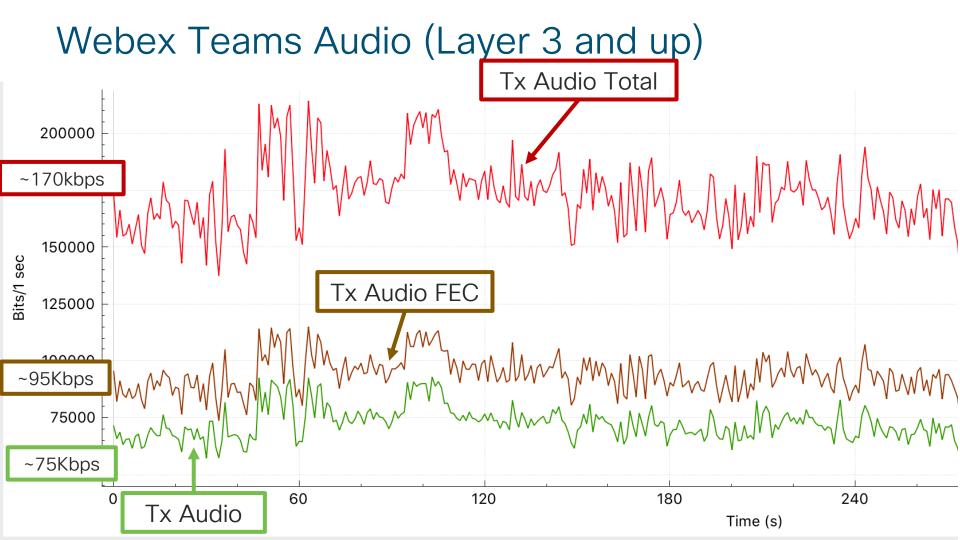
- Webex Mtgs App Desktop/Mobile: 1.8mb-2mb
 - Grid 5x5 = 5MB (Max)
 - High FPS Content Share: Addtl 1mb 2.5mb (max=intensive video)
- Teams Clients and Apps Desktop/Mobile: 2mb
 - Teams Clients: 2 Mbps at 720p resolution, 1 Mbps at 540p resolution and 500 kbps at 360p resolution
- Cloud Registered Endpoints (Single/Dual): 2mb/4mb
- Video Mesh Per Meeting Cascade: 12mb

Using these values is a good starting point, but only a starting point!



Your mileage may vary

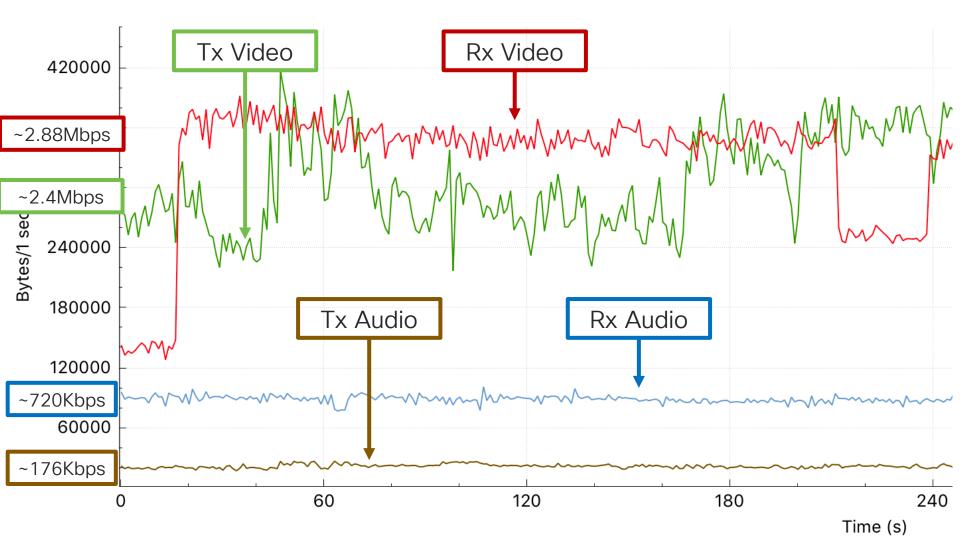


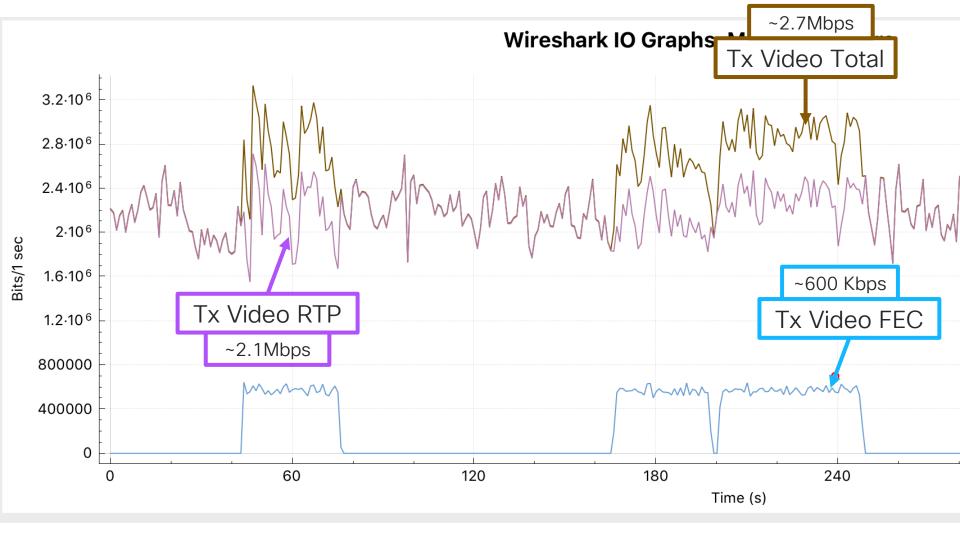


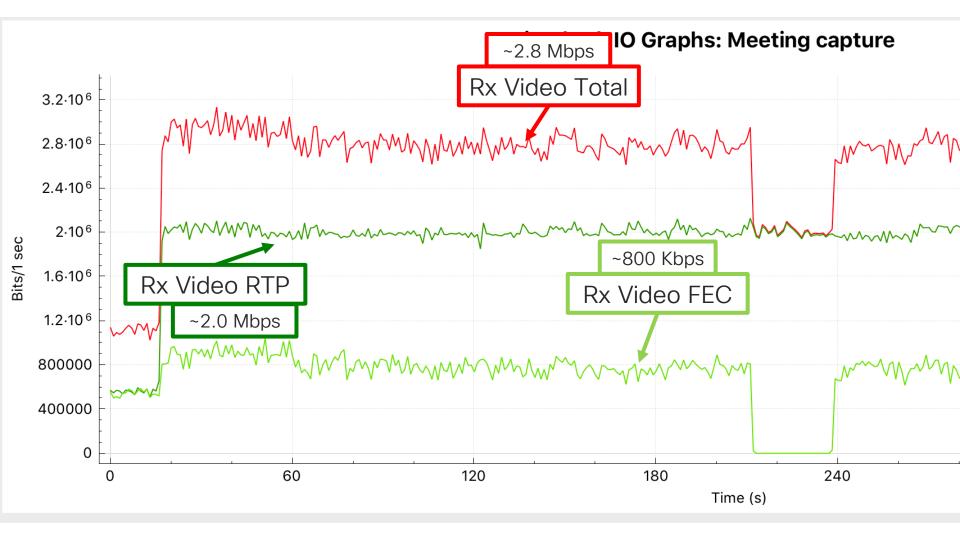
Approximate Audio bandwidth Utilization For Provisioning Purposes – Avg values

- Due to multistream audio more bandwidth is required for meetings with groups of 3 or more. We recommend provisioning approx 100–120k per user to account for the meeting bandwidth.
- Webex Teams Audio bandwidth uses about ~170k avg
 - 100% packet redundancy!
- Webex Meetings App Audio uses about ~70k avg
 - Packet redundancy increases as packet loss rises

Your mileage may vary





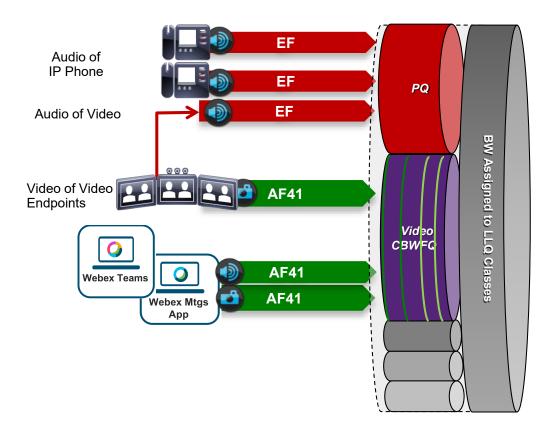


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Webex	Teams	

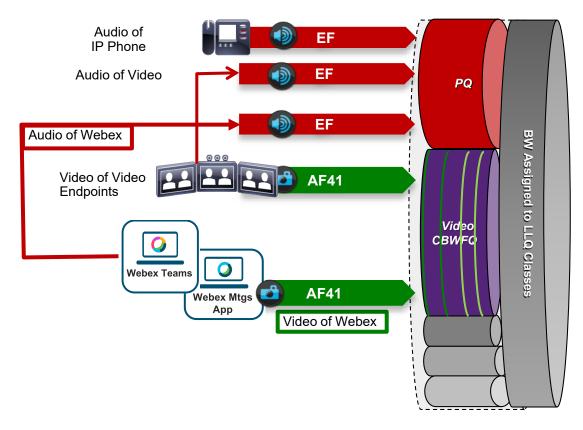
Vedex reams					
	↑ Transmit		\downarrow Receive		
Video	Main	Screen Share	ain	Screen Share	
Resolution	1280 x 720	/	1280 x 720	/	
Frame rate	29 fps	/	30 fps	/	
Bit rate	1557 kbps	/	1509 kbps	/	
Packet loss	0 %	0 %	0 %	0 %	
Roundtip/Latency	200 ms	0 ms			
Jitter	0 ms	0 ms	9 ms	0 ms	
Transport	UDP	UDP	UDP	UDP	
Audio					
Codec	Opus		Opus		
Bit rate	21 kbps		251 kbps		
Packet loss	0 %		0 %		
Roundtip/Latency	195 ms				
Jitter	0 ms		9 ms		
Transport	UDP		UDP		

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Mixed Environment



Mixed Environment



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Key Takeaways

- Create a Simplified Ingress Remarking Policy Use throughout the network
 - UDP Port ranges and/or NBAR2 to classify traffic
- Egress Queuing Policy
 - Single video queue for AF class traffic model is recommended
 - Consider multiple classes of Video AF41, AF42, AF43 (Dependent on ability to differentiate video through ports/NBAR)
- If you have both Webex Audio and constant bit-rate (CBR) Audio consider putting Webex Audio in the Video Queue (AF41)
 - Provision bandwidth for it
- Size, Provision, Monitor and Re-evaluate!

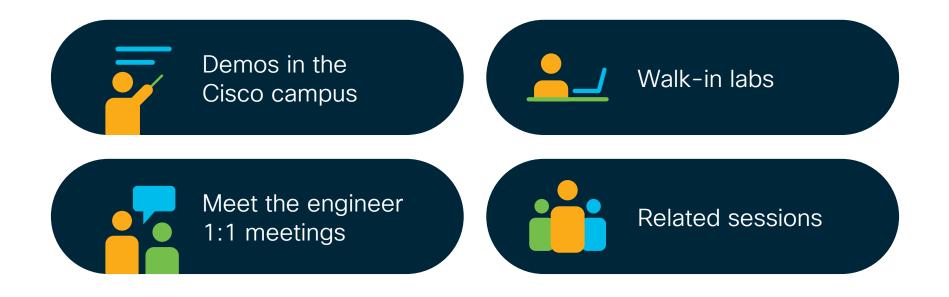
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- Please complete your session survey after each session. Your feedback is very important.
- Complete a minimum of 4 session surveys and the Overall Conference survey (starting on Thursday) to receive your Cisco Live t-shirt.
- All surveys can be taken in the Cisco Events Mobile App or by logging in to the Content Catalog on <u>ciscolive.com/emea</u>.

Cisco Live sessions will be available for viewing on demand after the event at <u>ciscolive.com</u>.

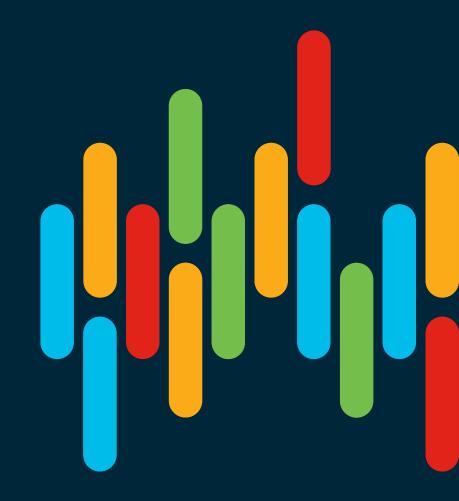
Continue your education



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Thank you



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