



T.38 Fax over IP Design Best Practices

David Hanes Customer Advanced Engineering

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- Today's featured expert is David Hanes
- Ask him questions now about T.38 Fax Over IP



David Hanes

Technical Leader in Customer Advanced Engineering, Cisco

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Expert Series





T.38 Fax over IP Design Best Practices

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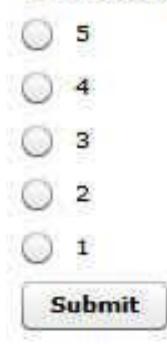
Limit one per attendee

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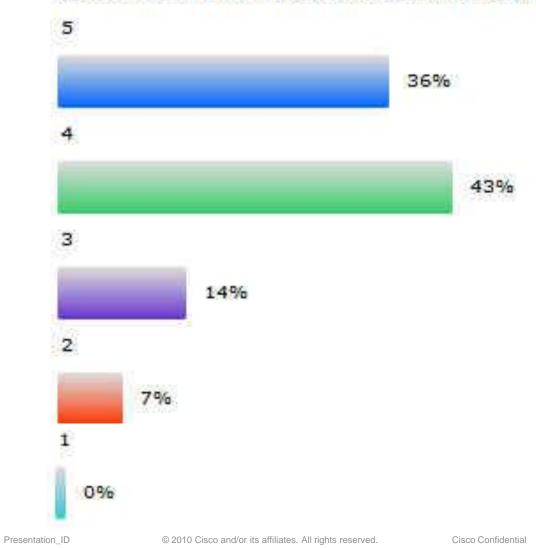
Poll Question

On a scale of 1 to 5 with 1 being "not important at all" and 5 being "very important or critical", please rate the importance of fax communications within your organization.



Poll Response

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Agenda

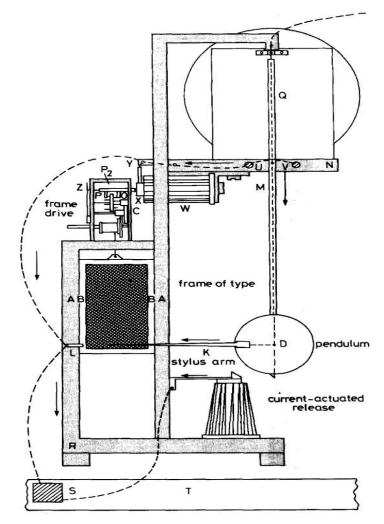
- Introduction to FoIP and T.38
- QoS Design Considerations
- Implementation Models
- Best Practices
- Key Takeaways



Introduction to FoIP and T.38

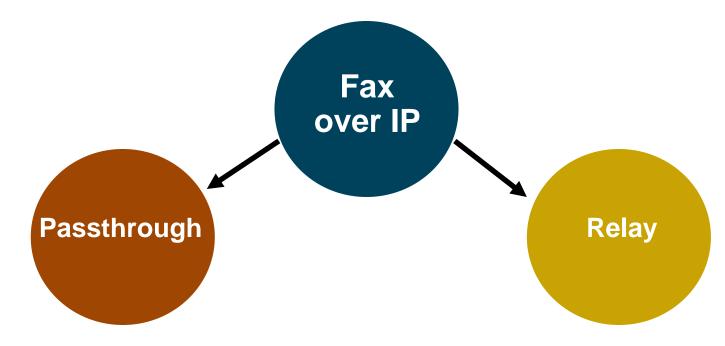
Fax Communications

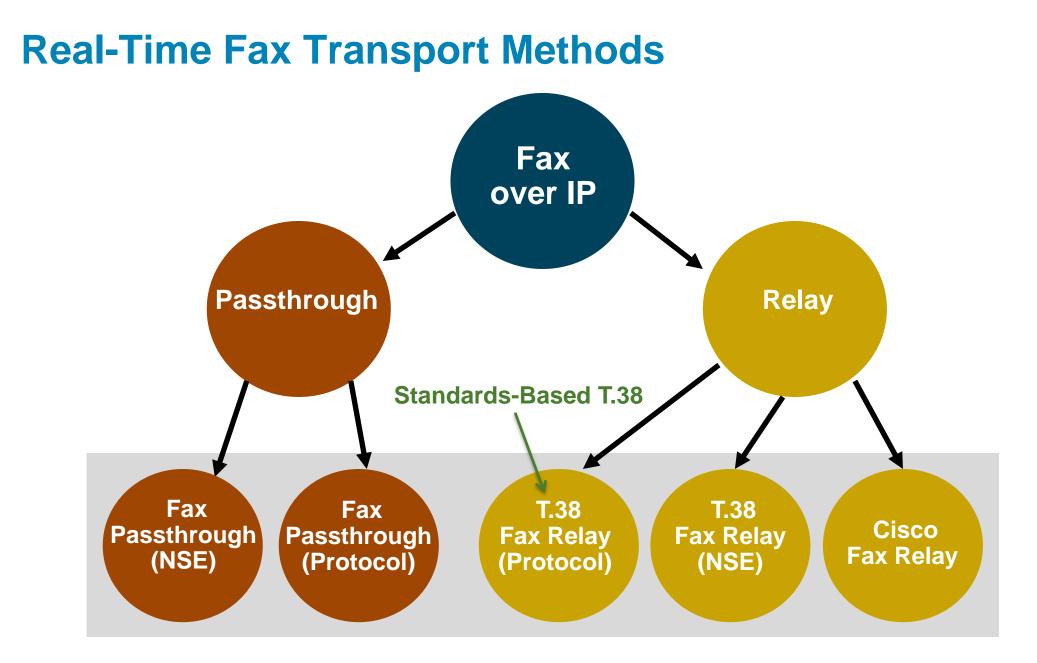
- Fax invented by Alexander Bain in 1843 (patented 30 years before the telephone)
- Fax is a ubiquitous form of communication today and Fax over IP (FoIP) is often overlooked in IP Telephony
- Group 3 (G3) is today's fax standard (speeds up to 14.4 kbps)
- Super G3 is an optional extension of G3 that allows for speeds up to 33.6 kbps



Alexander Bain's Facsimile Apparatus, British Patent 9745

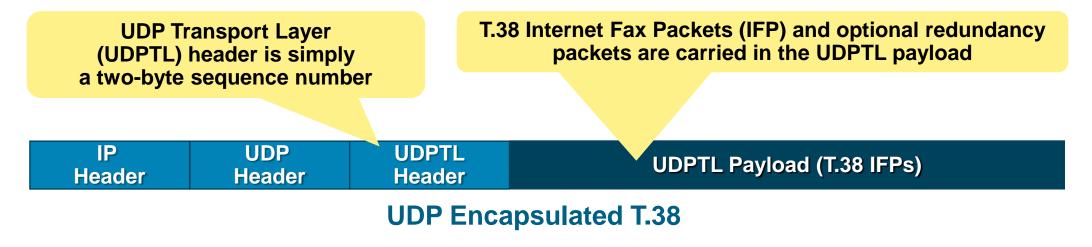
Real-Time Fax Transport Methods





T.38 Fax Relay

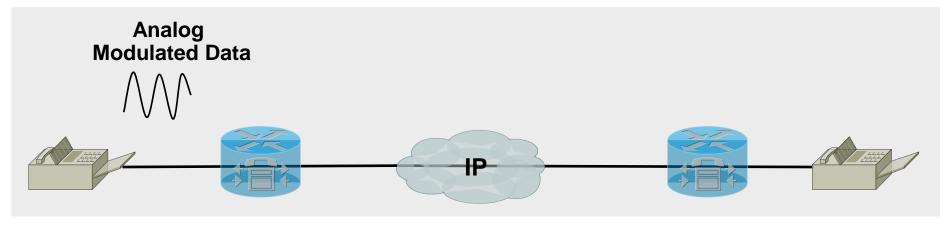
- T.38 is the de facto standard for handling fax transmissions today
- Cisco has always supported the 1998 version (known as version 0) of the ITU-T T.38 specification with UDP/UDPTL encapsulation
- In IOS version 15.1(1)T Cisco started supporting SG3 fax over T.38 (version 3)
- Cisco products can use either NSEs (proprietary) or the call control protocol (standards-based) to switch a call to T.38 mode
- Due to the absence of an RTP header, no support for SRTP and CRTP



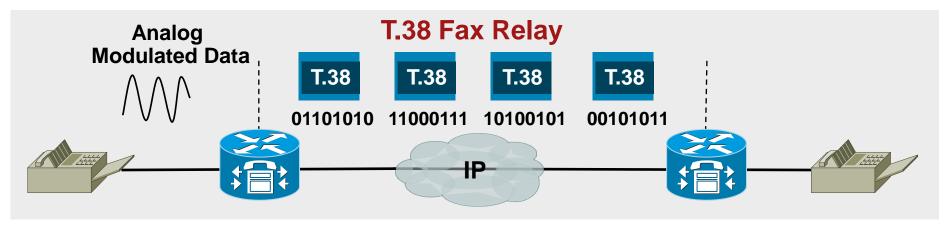
- The analog modulated fax data is demodulated by a Digital Signal Processor (DSP) on the gateway and the binary information is extracted
- Binary information is encapsulated in T.38 packets and then transported over the IP network
- A DSP on the destination gateway extracts the binary information carried within the T.38 packets and re-modulates it into a traditional analog fax signal on the telephony side



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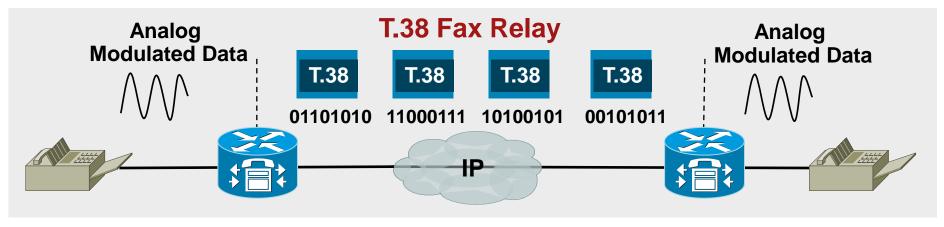


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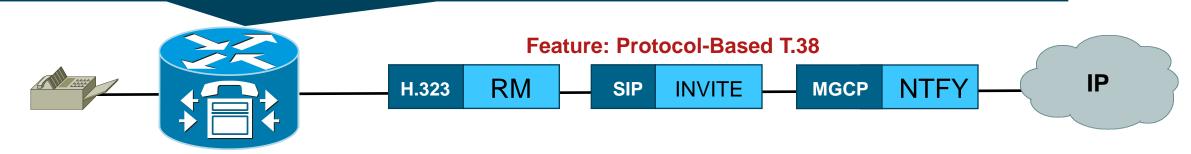


Configuring Standards-Based T.38

SIP/H.323:

dial-peer voice 1 voip fax protocol t38 version [0|3] Is-redundancy 0 hs-redundancy 0 fallback cisco

Note: Version 0 configures legacy T.38 G3 fax support, whereas version 3 enables the newly supported SG3 over T.38 feature [as of 15.1(1)T]

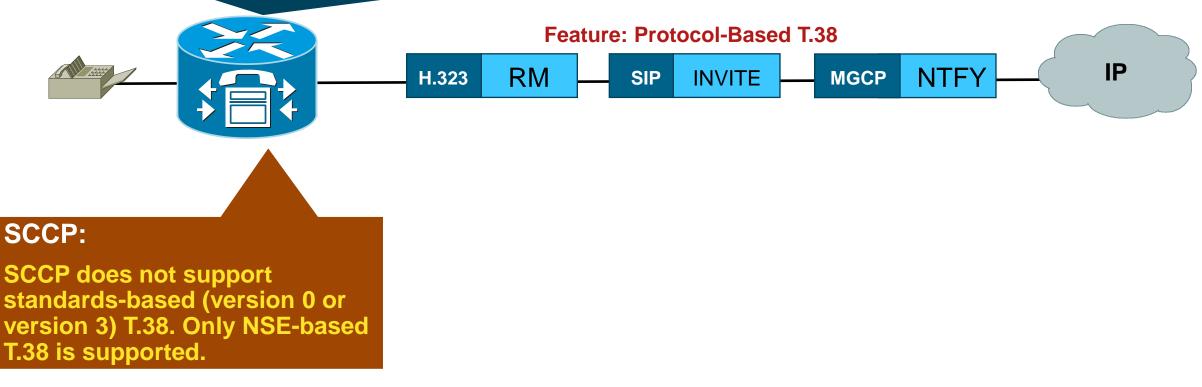


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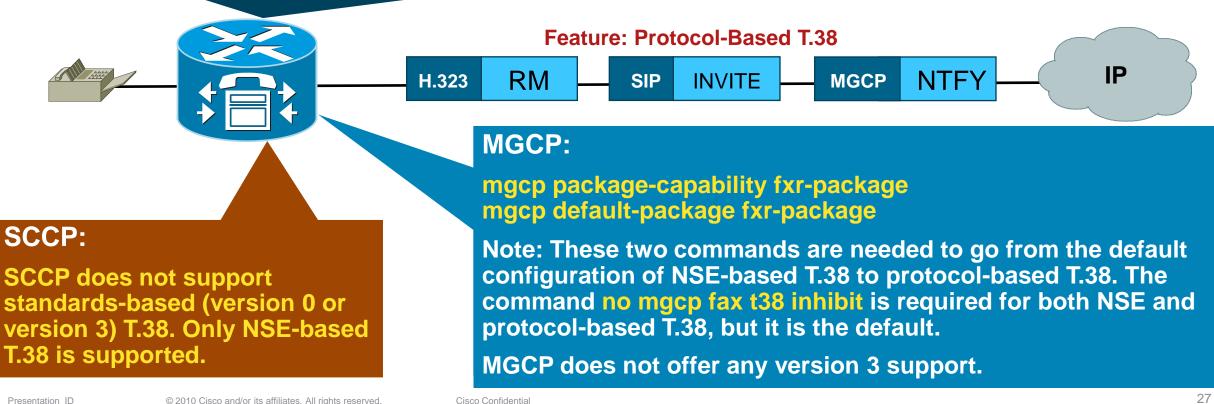


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Poll Question

Please mark the answer that best matches your organization's approach to the use of T.38 for fax communications.

- Have already deployed T.38
- In the process of rolling out T.38
- Considering the use of T.38
- Minimal or no interest in T.38

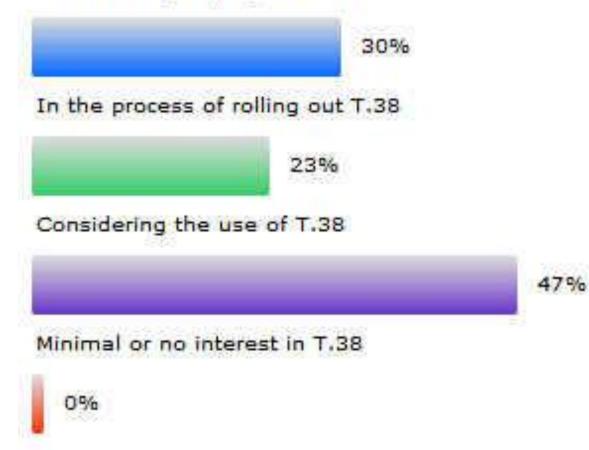


Poll Response

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Have already deployed T.38

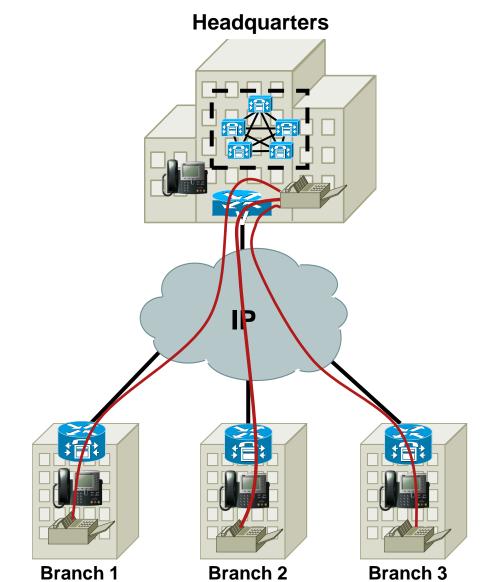




QoS Design Considerations

Are VolP QoS Policies Sufficient for Fax?

- In many cases, fax devices are located where VoIP is already installed
- You can "piggyback" fax traffic onto the same QoS policy implemented for VoIP
- As a rule of thumb, if good voice quality exists between locations, then fax communications should also work



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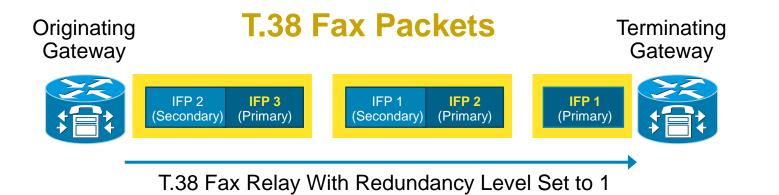
Originating Gateway



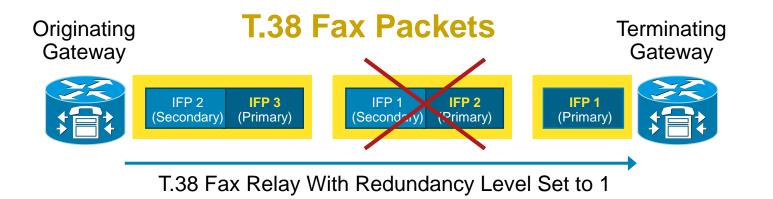
Terminating Gateway



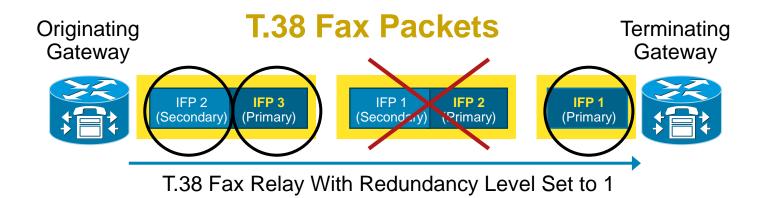
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- Ideally no packet loss should occur for a fax call
- If packet loss is present then enable T.38 with redundancy
- Enabling T.38 redundancy requires more bandwidth



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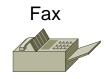


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QoS Design Parameters for T.38 Fax

	Delay	Jitter	Packet Loss
Voice	< 150 ms (one- way, mouth to ear)	< 30 ms (average, one- way)	< 1%
T.38 Fax	< 1000 ms	< 300 ms	None (unless T.38 with redundancy is enabled)

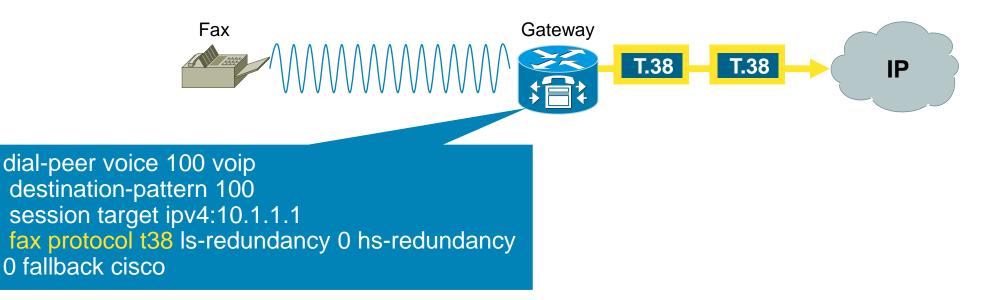
T.38 QoS Marking Configuration on Gateways





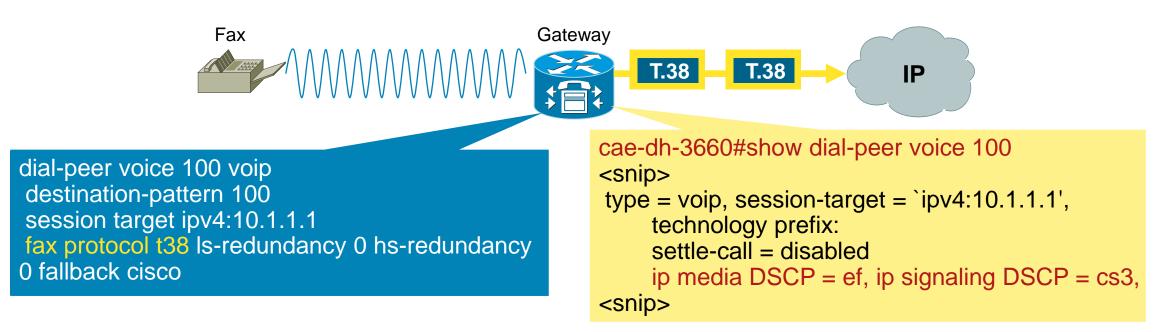


T.38 QoS Marking Configuration on Gateways



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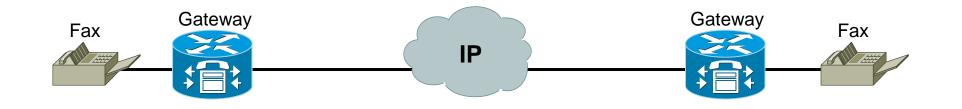
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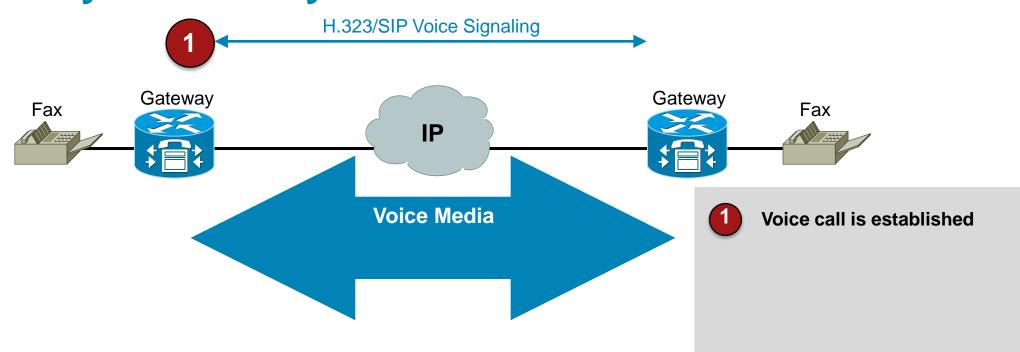
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- Dial-peer QoS markings can be viewed with the command show dial-peer voice
- Default dial-peer QoS markings are cs3 for signaling and ef for media



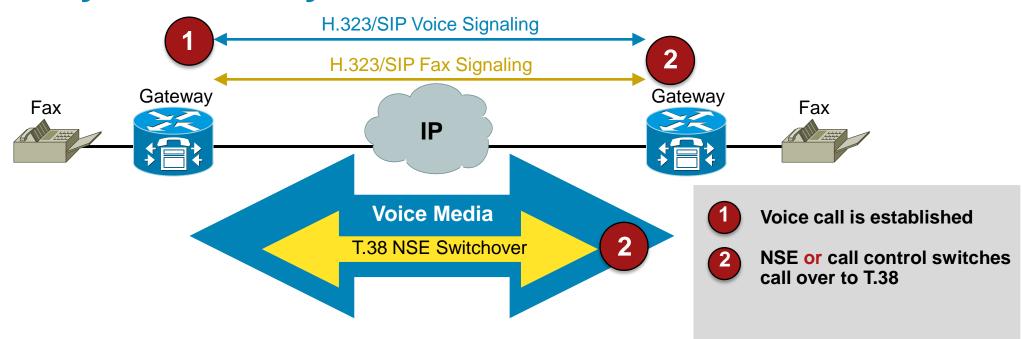
Implementation Models



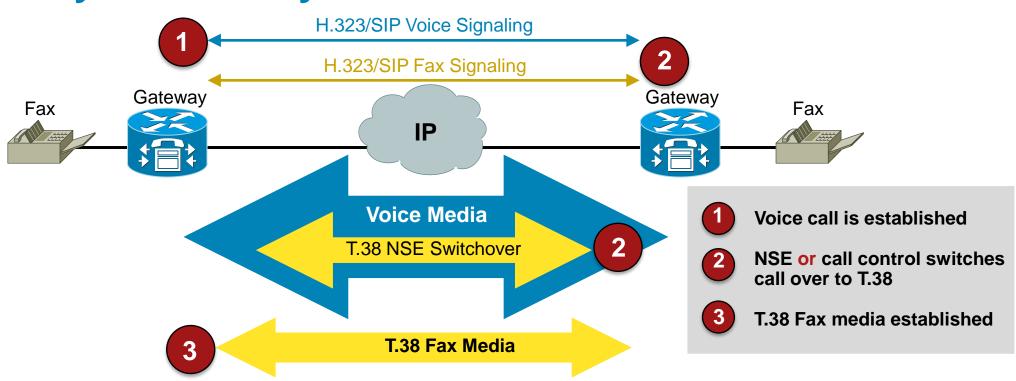
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- Signaling and media packets are exchanged between the voice gateways
- T.38 NSE-based switchovers occur within the voice media stream while T.38 protocolbased (standards-based) switchovers happen within the call control protocol



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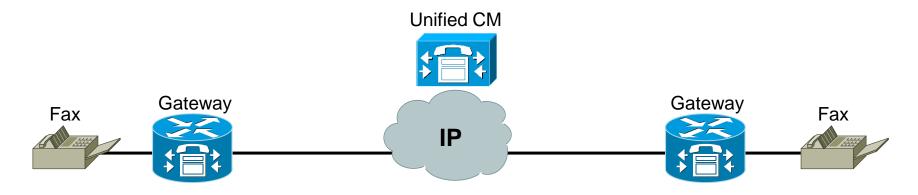


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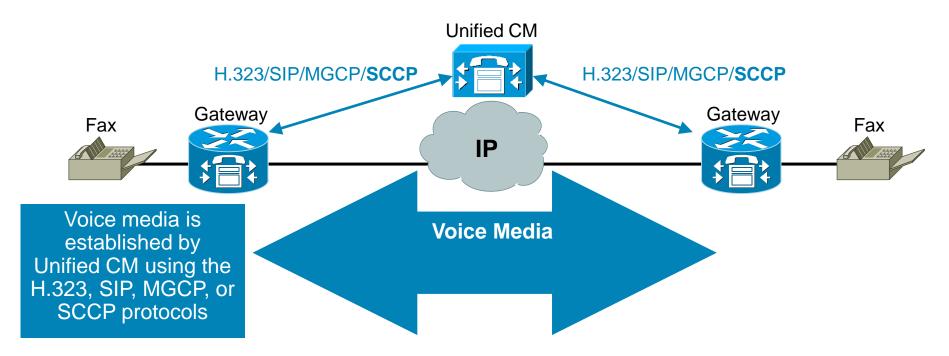
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Cisco Unified Communications Manager, Gateway-Controlled



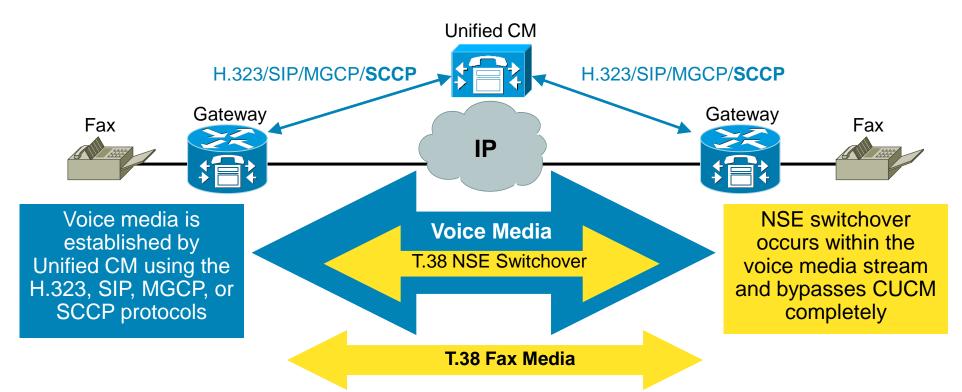
- Cisco Unified Communications Manager (CUCM) controls the setup of the initial voice call between the gateways using standard call control signaling such as H.323, SIP, MGCP, and SCCP
- Skinny (SCCP) gateways only support this integration model

Cisco Unified Communications Manager, Gateway-Controlled



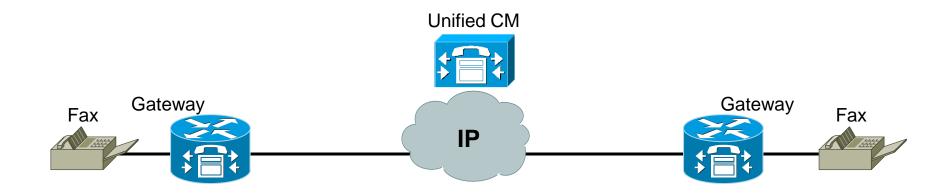
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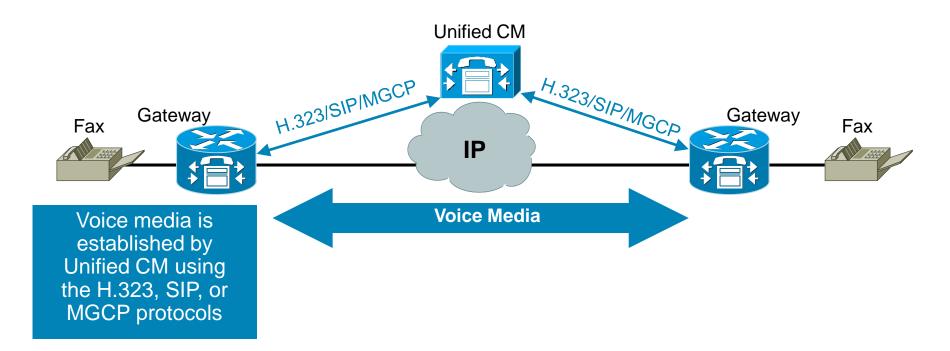
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Cisco Unified Communications Manager, CA-Controlled



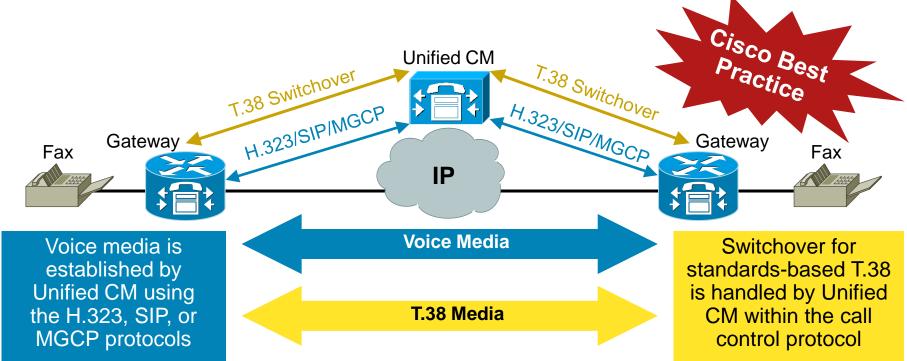
- Cisco Unified Communications Manager controls the setup of the initial voice call between the voice gateways and controls the switchover to T.38 fax relay
- Instead of the gateways independently handling the switchover, Cisco Unified Communications Manager coordinates the switchover using the call control protocol

Cisco Unified Communications Manager, CA-Controlled



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Cisco Unified Communications Manager and T.38 Support

Call Control Protocol Support for T.38	Cisco Unified Communications Manager Software Release	
H.323	4.1(1), 4.2(3), 5.0(1), and 6.0(1)	
H.323 and MGCP	4.2(3) and 6.0(1)	
H.323 and SIP	5.0(1) and 6.0(1)	
H.323, SIP, and MGCP	6.0(1) and later	

- Only release 6.0(1) and later offers support for T.38 within the call control protocols of H.323, SIP, and MGCP
- The SCCP protocol only handles T.38 using an NSE-based switchover



Best Practices

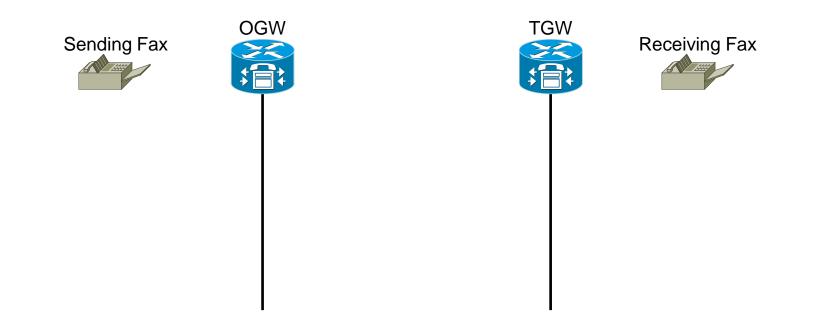
FoIP Bandwidth Utilization

- T.38 offers the best bandwidth savings compared to other FoIP transports
- For CAC (Call Admission Control) planning, T.38 without redundancy is roughly equivalent to G.729
- Be cautious when enabling T.38 redundancy to ensure that CAC thresholds are not exceeded

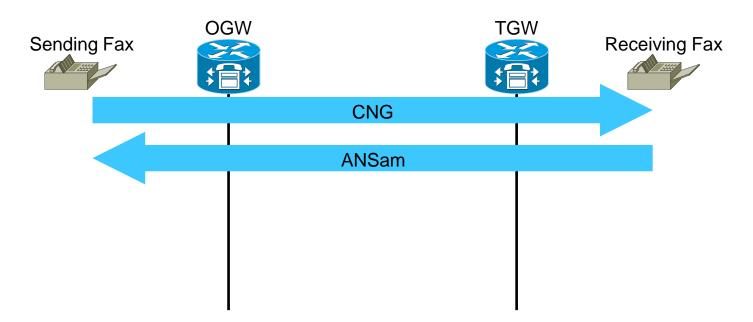
¹Values are approximate with Ethernet or Frame Relay headers

²Values are peak and only occur unidirectionally during the sending of a page at 14.4 Kbps

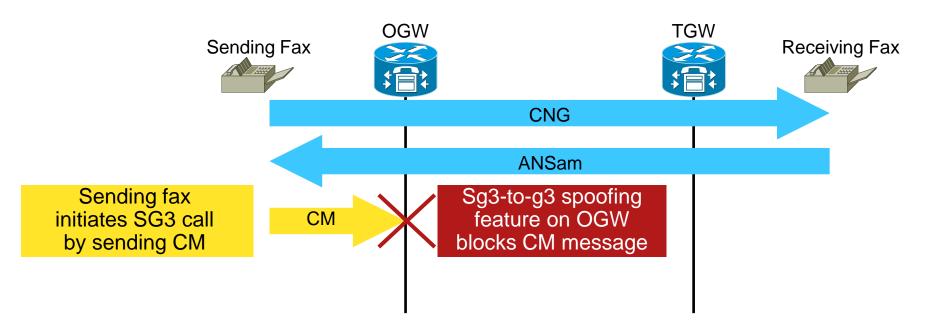
Codec	Bandwidth ¹
G.711 (64 Kbps)	83 Kbps
G.729 (8 Kbps)	27 Kbps
G.723 (6.3 Kbps)	19 Kbps
Fax passthrough/ passthrough (G.711)	83 Kbps
Fax passthrough (G.711) with redundancy	170 Kbps
T.38 (no redundancy)	25 Kbps ²
T.38 (redundancy level 1)	41 Kbps ²
T.38 (redundancy level 2)	57 Kbps ²
Cisco fax relay	48 Kbps ²



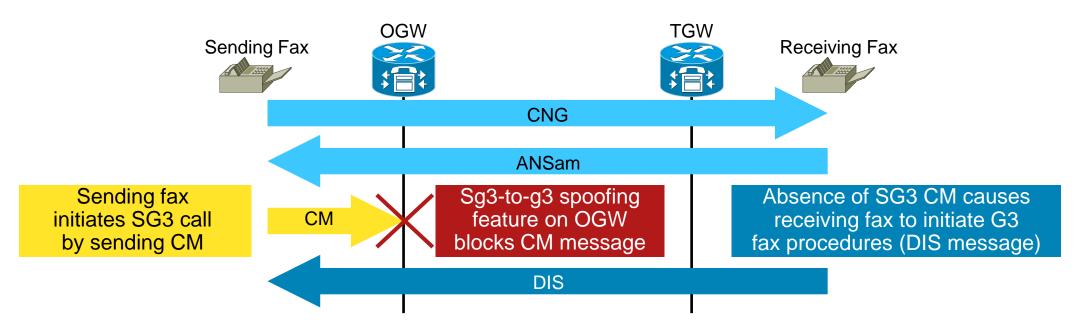
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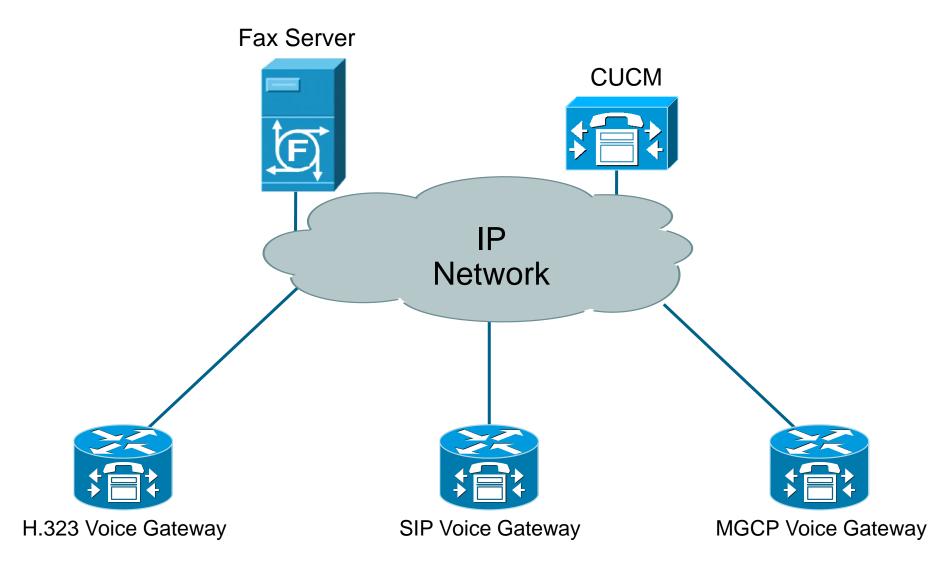


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SIP Trunks and FoIP

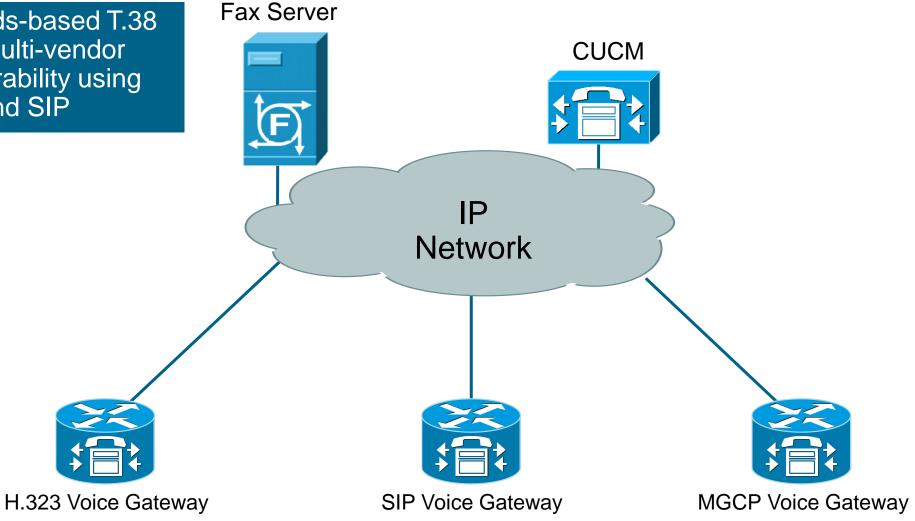
- The three common fax transport methods available with SIP trunking are T.38, G.711 voice, and fax passthrough
- T.38 is the best choice but it is still not offered by some providers and interoperability problems are still being resolved in some cases

Fax Method	T.38 Fax Relay	G.711 Voice	Passthrough
Pros	 Highest fax success rates can be achieved Cleanest solution from signaling and media point of view Use less bandwidth than G.711 	 Most widely deployed Simplest solution 	 Provides bandwidth savings as G.729 voice call only upspeeds to G.711 if call is fax
Cons	 Interoperability between different vendors Not offered by many Service Providers 	 Consumes a large amount of bandwidth for all calls Sensitive to impairments, no redundancy 	 Interoperability between different vendors Not supported by Cisco Unified Communications Manager Sensitive to impairments, no redundancy



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Standards-based T.38 allows multi-vendor interoperability using H.323 and SIP



Fax Server Standards-based T.38 allows multi-vendor CUCM interoperability using H.323 and SIP H.323/SIP F 323 80°. Network H.323 Voice Gateway SIP Voice Gateway MGCP Voice Gateway

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Fax Server Standards-based T.38 allows multi-vendor CUCM interoperability using H.323/SIP H.323 and SIP H.323/SIP F 323 80°. Sib Network MGCP Voice Gateway H.323 Voice Gateway SIP Voice Gateway

Fax Server Standards-based T.38 Standards-based T.38 H.323/SIP and CUCM allows fax allows multi-vendor CUCM interoperability using server access to MGCP H.323/SIP H.323 and SIP voice gateways H.323/SIP 1.38 MGCP 80°. Sib 1.38 Network H.323 Voice Gateway SIP Voice Gateway MGCP Voice Gateway

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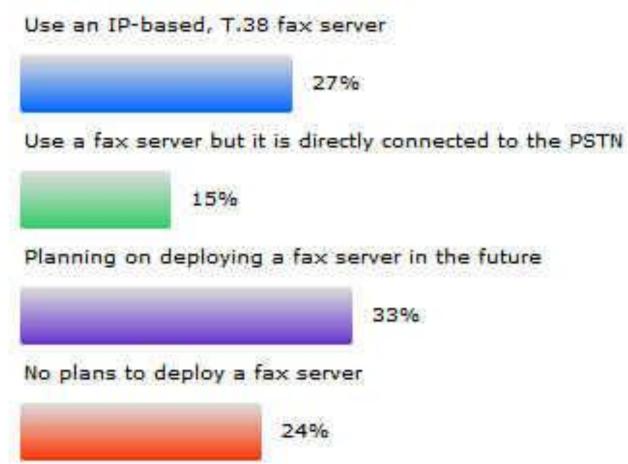
Poll Question

Please mark the answer that best describes your organization's use of a fax server?

- Use an IP-based, T.38 fax server
- Use a fax server but it is directly connected to the PSTN
- Planning on deploying a fax server in the future
- No plans to deploy a fax server

Submit

Please mark the answer that best describes your organization's use of a fax server?





Key Takeaways

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1. Cisco gateway products offer T.38 with either a proprietary NSE-based switchover or a standards-based switchover in the call control protocol. Standards-based T.38 should always be your first choice.



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4. T.38 supports SG3 in IOS version 15.1(1)T and later for SIP and H.323 only. No T.38 SG3 support for MGCP and SCCP voice gateways.

5. T.38 is recommended over SIP trunks but be aware that not all service providers support T.38 yet and interoperability issues are still being addressed in some cases.

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8. T.38 uses a UDPTL header instead of RTP so secure RTP (SRTP) and compressed RTP (CRTP) are not available.



Q&A

Post Questions on Our Forum Here:

https://supportforums.cisco.com/community/netpro/ask-the-expert



Join us for our next "Ask The Expert" webinar!

Topic: AnyConnect Configuration and Troubleshooting

December 13th 11pm PT / December 14th 2am ET / New Delhi 12:30pm

Register at: http://www.ciscolive.com/ate



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Topic: Introduction to CiscoWorks LMS 4.0

December 1, 10-11 a.m. Pacific Time

Join us for a view into the new CiscoWorks LMS 4.0. This engineers session is going to be packed with demos, comparisons between this and other products and tips/tricks to finally get a LAN Management System that works for you.



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