

SIP Conferencing

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Conference International SIP – Paris, France

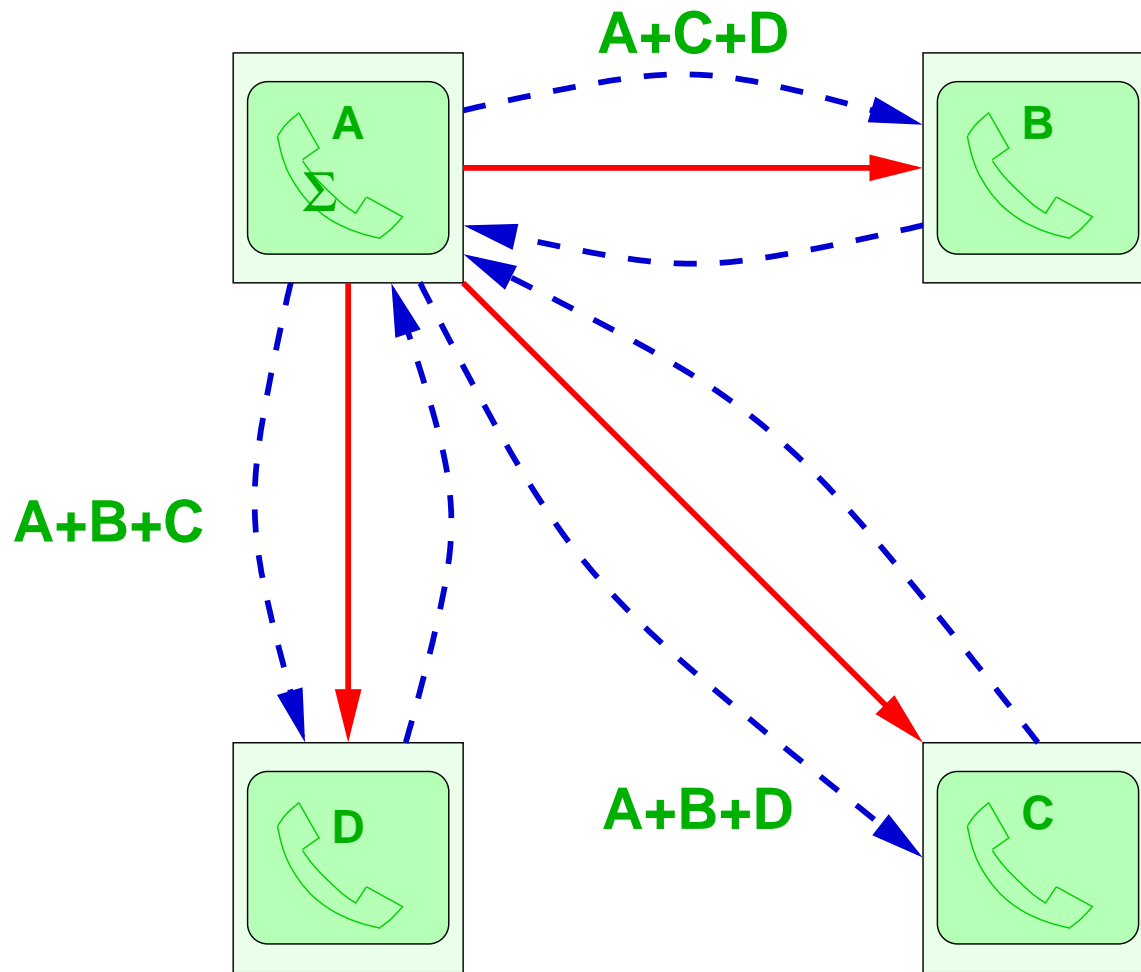
February 21, 2001

Joint work with Gautam Nair, Jonathan Rosenberg and Kundan Singh

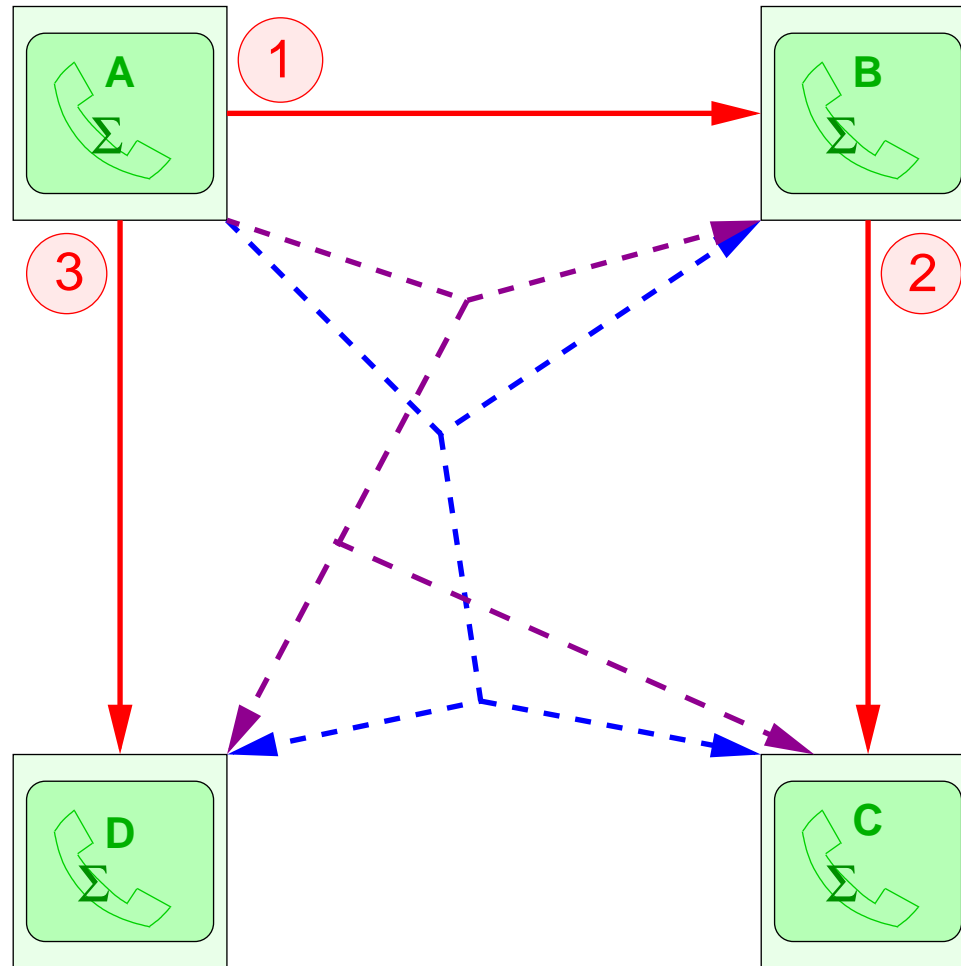
Overview

- conferencing models
- full-mesh conferencing
- implementation of a conferencing server
- open issues

Conference models: end system mixing

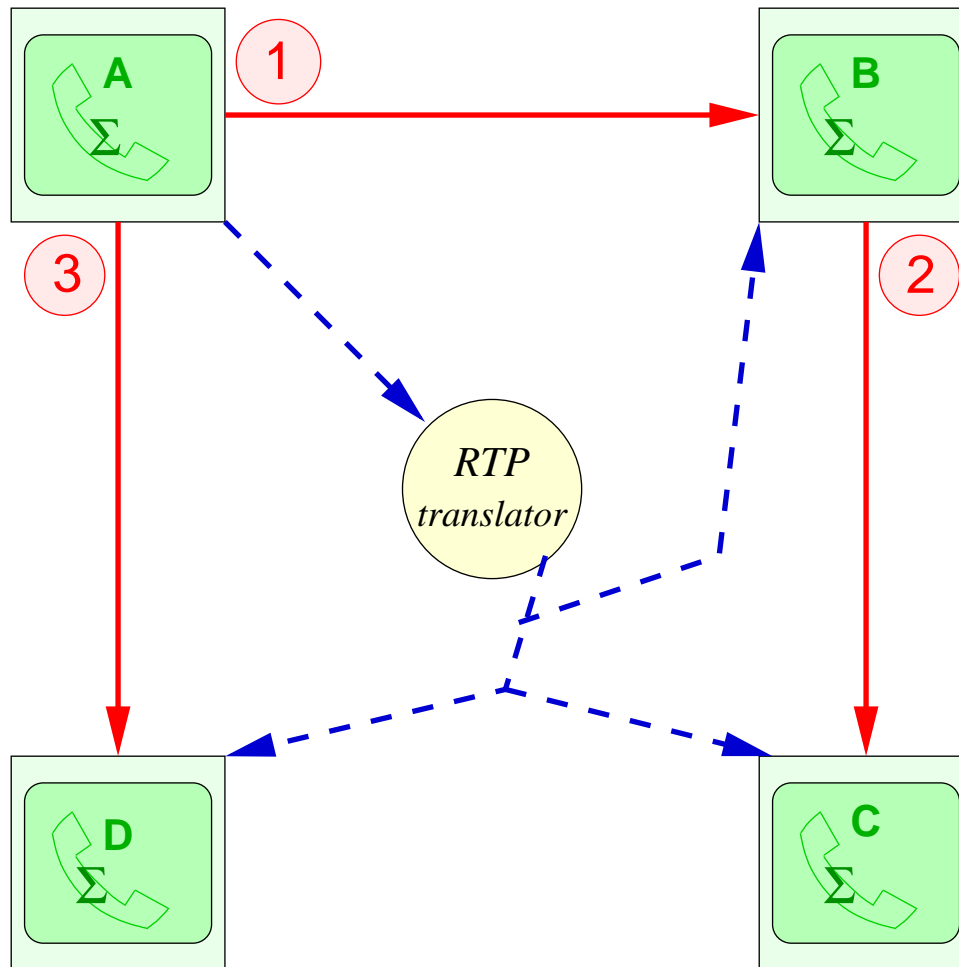


Conference models: multicast transmit & receive



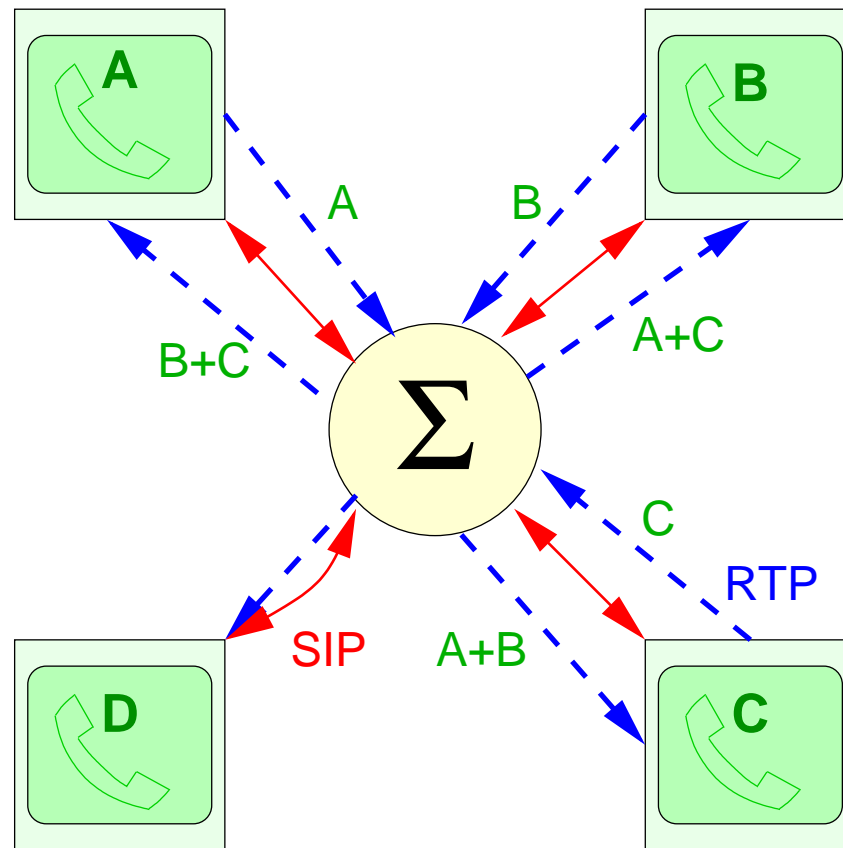
May be suitable for small-scale multicast proposals (“xcast”)

Conference models: multicast receive, unicast transmit



Conference models: central server

can be call-out or dial-in

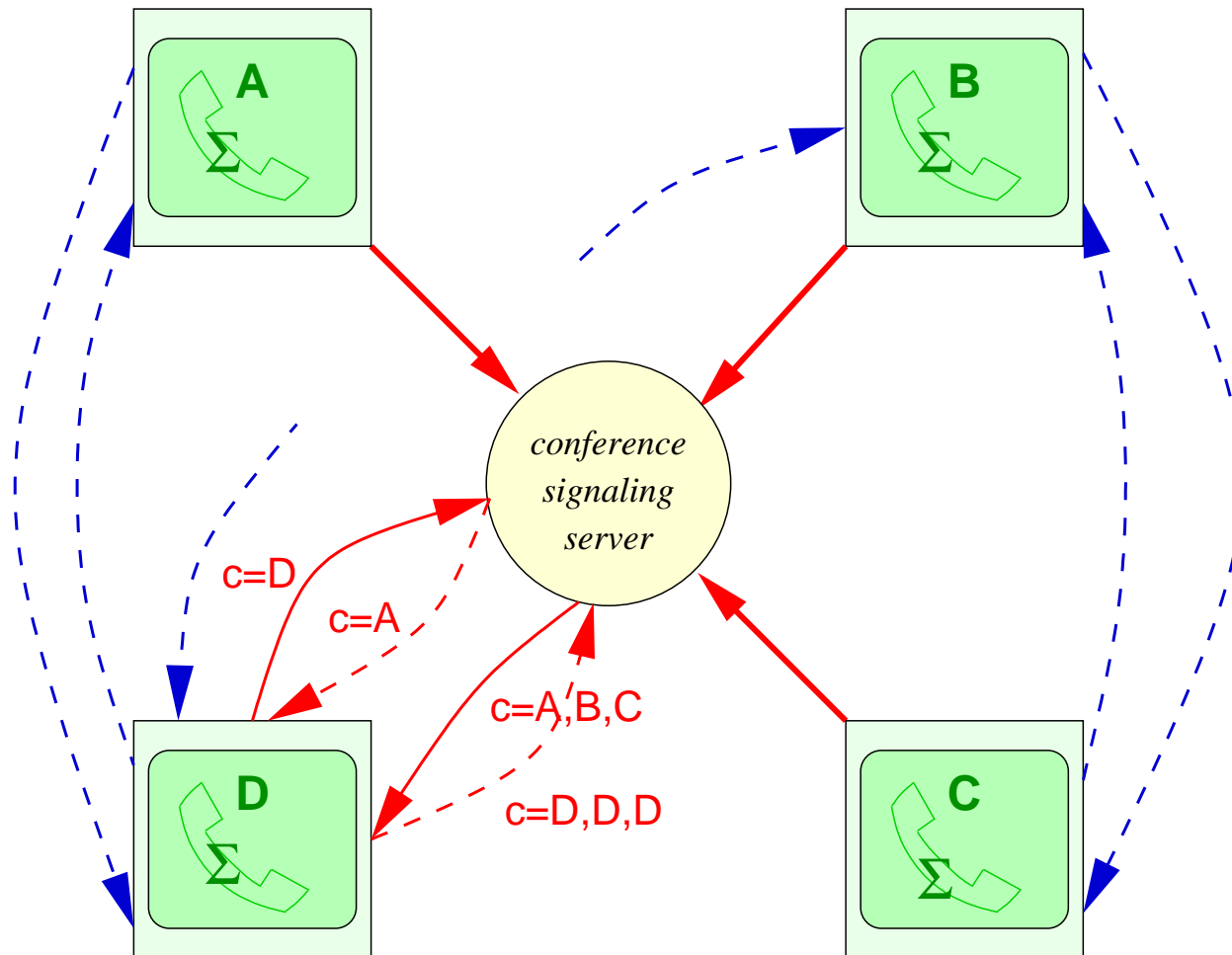


Conference models: central server

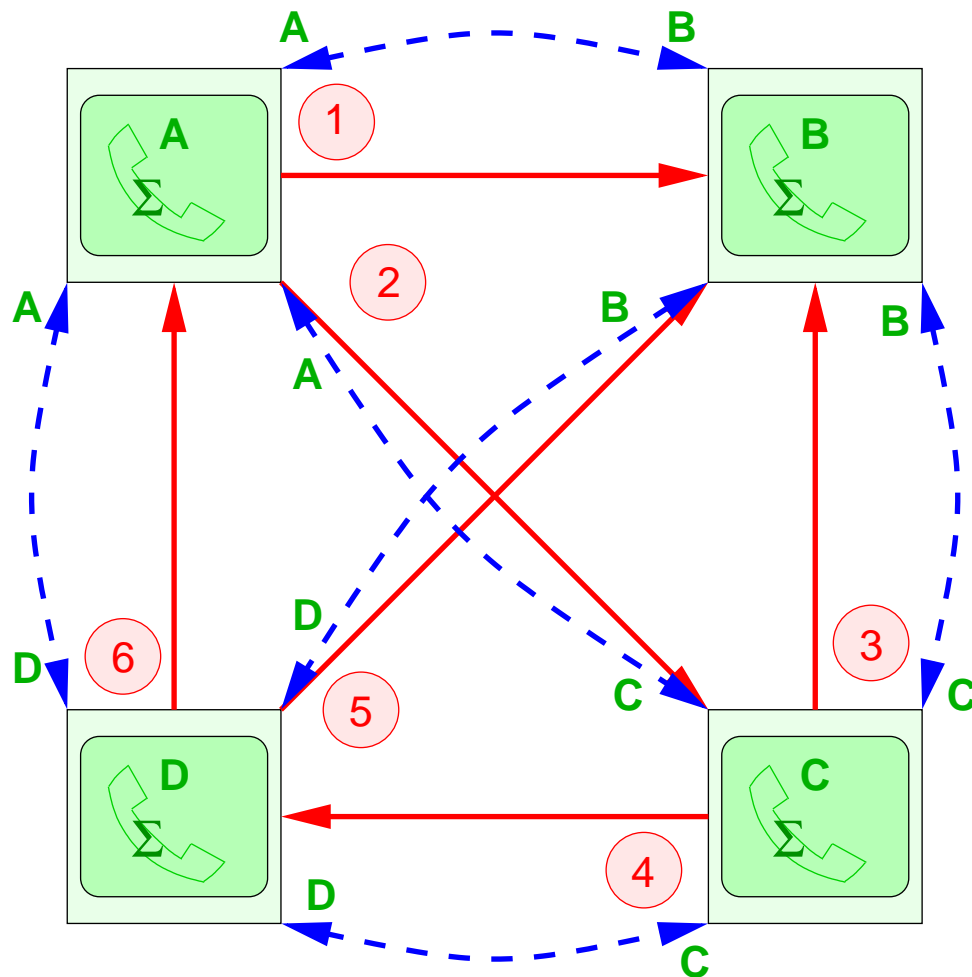
- conference identified by SIP URL, e.g., sip2001@upperside.fr
- ad-hoc conferences useful for three-party calls
- can create ad-hoc conference by using random URL, e.g., sip:a7hytaskp09878a@a.servers.com, use REFER to get others to add themselves to that conference
- for ad-hoc, conference lasts until last one leaves

Conference models: decentralized

Central signaling server, mesh media



Conference models: full mesh



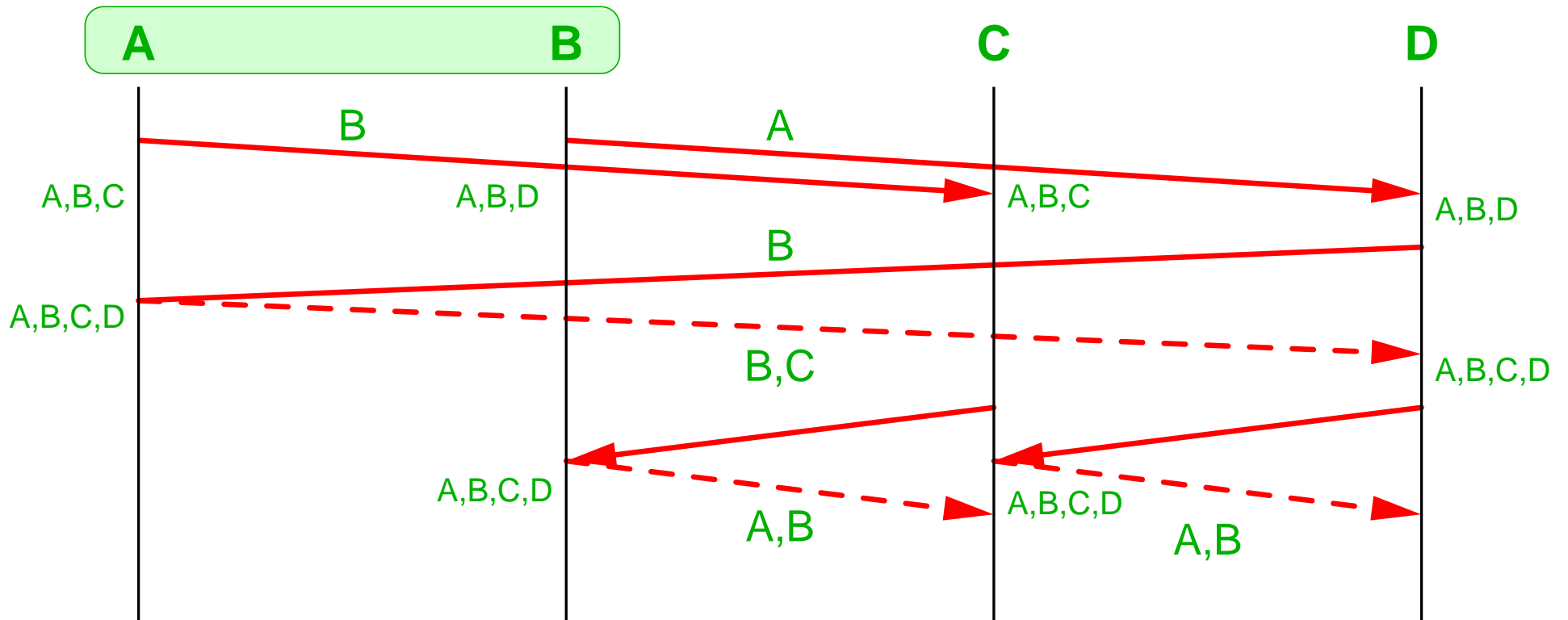
A invites B
 A invites C
 C invites B
 C invites D
 D invites A, B

Full mesh

- +: all participants are equal
- +: easy to control media distribution
- need to deal with simultaneous call-outs of two parties
- departures while other party joins
- UAC and UAS send membership list in request and 200
- if either learns about a new member, send INVITE
- allows joining of separated clouds
- departing member sends BYE to everyone – may need to refuse INVITEs from new arrivals (partitions!)

Conference models: full mesh

A invites C and B invites D *simultaneously*:



Full mesh: open issues

- conference identification?
- how to identify that call legs are related (should be mixed)
- call-in for mesh?
- refresh mechanism to heal network partitions

Conference models – a comparison

Name	SIP	media	adding	joining	members	scale
End-Mixing	tree	tree	normal INVITE	normal INVITE	RTCP	small
Multicast	pairs	mcast	normal INVITE	mcast join	RTCP	large
Dial-In	star	star	REFER	normal INVITE	RTCP	medium
Ad-Hoc	star	star	REFER	normal INVITE	RTCP	medium
Dial-Out	star	star	REFER	normal INVITE	RTCP	medium
Decentral	star	mesh	refer +	normal	RTCP	medium
Mesh	mesh	mesh	INVITE	?	SIP	medium

Conference models – complexity

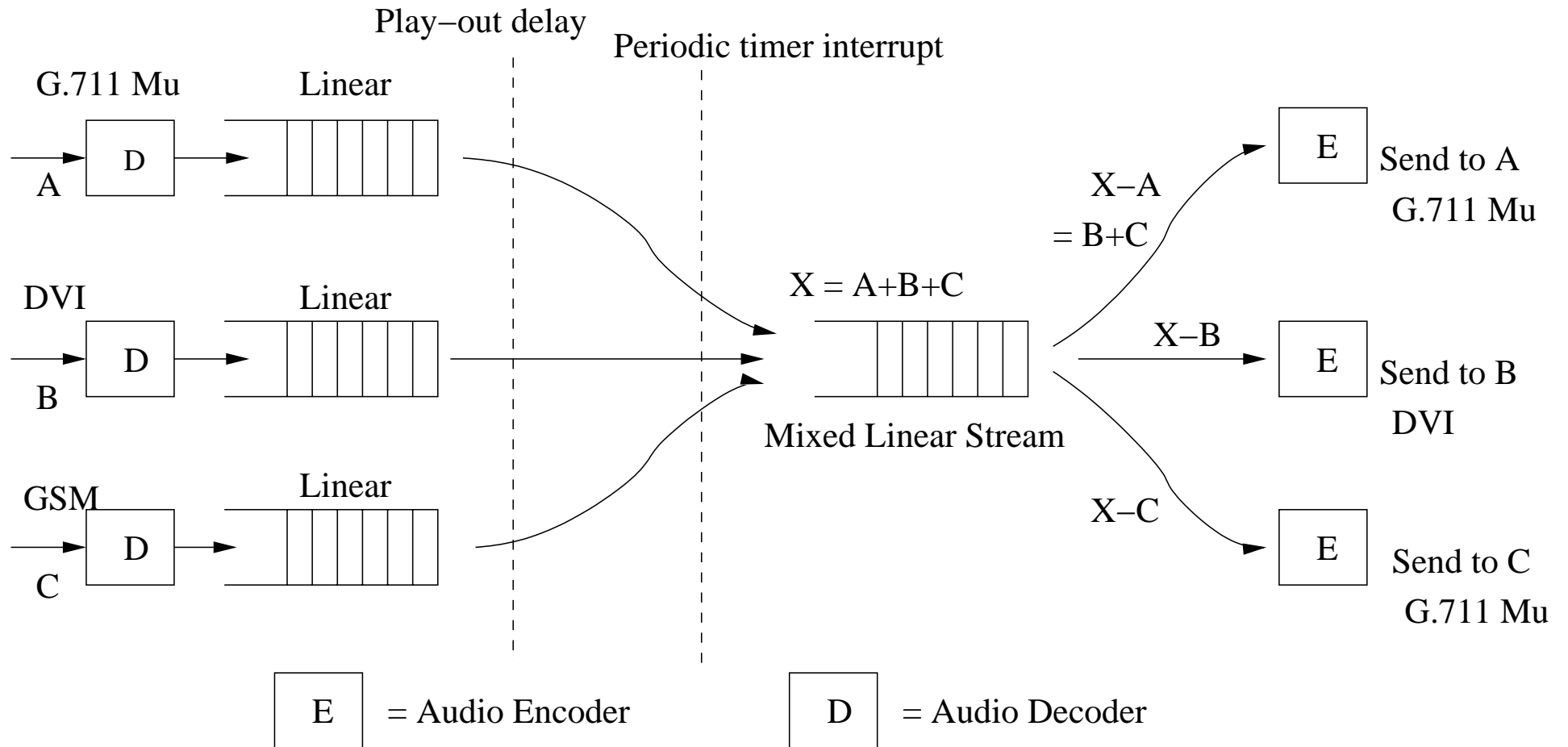
M active senders, N participants

Properties	central	full mesh	mcast	uni rx, mcast tx	end mixing
Topology	Star	full mesh	m-cast tree	star + mcast tree	ad-hoc
Server proc.	$O(M + N)$	n/a	n/a	$O(M + N)$	n/a
Endpoint proc.	$O(1)$	$O(M)$	$O(M)$	$O(1)$	variable
Server bw	$O(M + N)$	n/a	n/a	$O(M)$	n/a
Endpoint bw	$O(1)$	$O(M)$	$O(1)$	$O(1)$	variable
Scaling	medium	medium	large	large	medium
Heterogen. UA	yes	yes	no	no	yes (partially)
Own media?	no	no	no	yes	no

Columbia sipconf conferencing server

- central-server model for mixing
- commodity hardware, easily scalable to rack full (conferences: avg. 5, 45 minute duration)
- mixes audio streams, replicates RTP (and UDP) streams
- works for video and text chat (RFC 2793)

Mixing heterogeneous streams



Example: Columbia software conference server

sipconf

Currently following conferences exist

Conference	Description
<input type="checkbox"/> irtmeeting@cs.columbia.edu	IRT group meeting.
<input type="checkbox"/> test	No description
<input type="checkbox"/> swisscom	Swisscom test
<input type="checkbox"/> 9998	Test
<input type="checkbox"/> aks	My conference

Enter password to [Delete](#) selected conference

You can create a new conference by giving the conference name. Conference name is of the form 'confname@host' or just 'confname' Do not forget to specify an easy to remember password which you can use for deleting the conference entry. You must specify a brief description of the conference.

Create new conference: [Create](#)

Description of the conference :

Your password, needed for deleting:

[Report a problem](#)
[What is sipconf? Help!](#)

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Performance for single conference

SPARC Ultra 10 with 350 MHz CPU:

Participants	CPU	memory	bandwidth (Mb/s)	
	(%)	(MB)	inbound	outbound
2	< 0.1	2.7	0.08	0.07
20	< 1	6.0	0.08	1.37
40	2-3	9.6	0.08	2.81
60	5	13	0.08	4.25
80	10-15	17	0.08	5.69
100	35-50	22	0.08	7.13
120	50-70	26	0.08	8.59

Performance for three-party conferences

Good quality up to 15 conferences

Confer- ences	partici- pants	CPU (%)	memory (MB)	bandwidth (Mb/s)	
				inbound	outbound
3	9	< 0.4	4.1	0.72	0.65
6	18	< 2.0	5.7	1.44	1.30
9	27	7-13	7.3	2.16	1.94
12	36	15-20	9	2.88	2.60
15	45	25	10	3.60	3.24
18	54	30	12	4.32	3.89

Conclusion

- need many different models of conferencing
- trade-off infrastructure vs. complexity vs. scaling
- mostly, can be handled by existing SIP mechanisms
- full mesh requires further work
- transition between conferencing models?