

SIP Conferencing

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IETF Conferencing

- ▶ **Packet multimedia experiments since the 1980s**
 - Audio/video tools + protocols for A/V over IP
 - Conference announcement and control protocols
- ▶ **First IETF Audiocast (1992)**
 - Mbone-based audio transmission from selected IETF working groups
- ▶ **Since then: IETF sessions on the Mbone**
 - Audio + video (+ sometimes slides)
 - Enabling remote participation (even talks)
- ▶ **Other uses of Mbone conferencing**
 - Broadcasting NASA missions, concerts, ...
 - Lectures, seminars, project meetings, ...

Traditional IETF Conferencing Concept

- ▶ **Multicast-based**
- ▶ **Loosely-coupled conferences**
 - no membership control
 - inexact information about participants
 - provided on a voluntary basis
 - security by encryption
- ▶ **Public announcements and invitations**
 - Convey session parameters, then get out of the way
 - Session Announcement Protocol (SAP)
 - Session Initiation Protocol (SIP)
- ▶ **Conference control**
 - Some need perceived; several attempts; insufficient real interest

Some Historic (IETF) Conferencing Protocols...

- ▶ **MMCC / CCP (early 1990s)**
 - Control for a distributed packet-based conferencing system
- ▶ **Agreement Protocol (1995)**
 - State synchronization in multicast groups
- ▶ **Conference Control Channel Protocol (CCCP) (1995)**
 - Communicating state and events between local and remote entities
 - Support for modular systems (media engines, controllers, UIs, ...)
- ▶ **SCCP (1996; revision 2001)**
 - Distribution of conference + media session state
 - Makes use of reliable multicast
- ▶ **Some other protocols...**

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All protocols were based on IP Multicast!

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SIP and Conferencing over Time...

- ▶ **Origin: MMUSIC: Multiparty Multimedia Session Control**
- ▶ **From Invitation... to initiation, modification, and termination**
- ▶ **From Multiparty... to point-to-point-focused**
- ▶ **From Multimedia... to voice-centric**

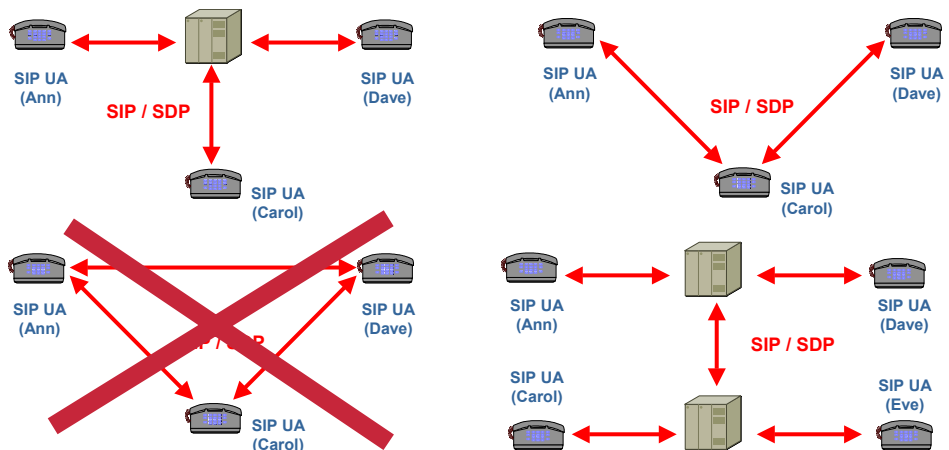
Now: Multiparty & multimedia rediscovered

But: Don't believe in multicast (anymore)!

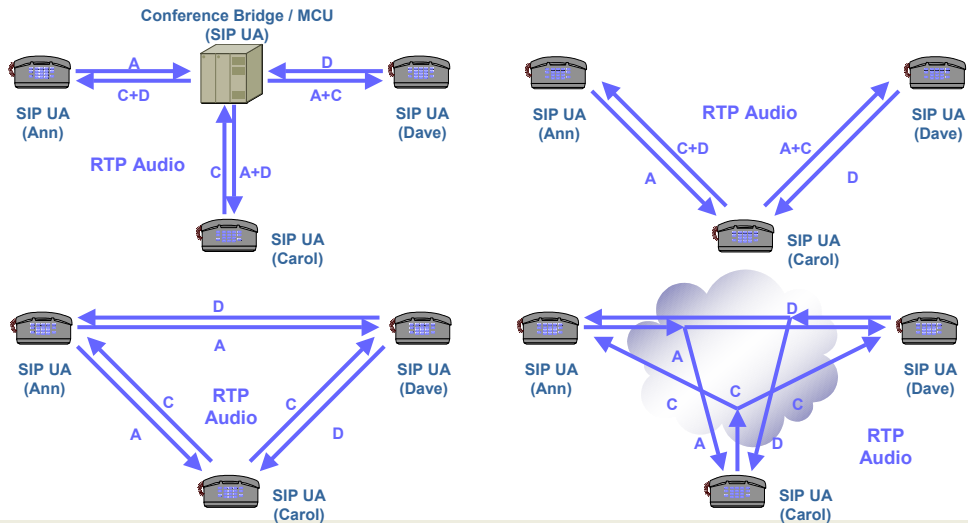
“More recent” Conferencing Models (2000/2001)

- ▶ SIP signaling relationships
 - Centralized (bridge, endpoint) vs. mesh of SIP dialogs
- ▶ Media distribution
 - Unicast vs. multicast
- ▶ Media mixing
 - Centralized (bridge, endpoint) vs. decentralized
- ▶ Conference creation
 - ad-hoc vs. scheduled
 - “dial-in” vs. “dial-out”
- ▶ How much functionality to provide in SIP?
 - Join / leave
 - Membership management? ...?

SIP Signaling Relationships



(RTP) Media Sessions



Current SIP Conferencing Framework

- ▶ **Motivators: n-way calling, video conferencing, ...**
 - Tightly coupled conferences
- ▶ **Recent conferencing models preserved**
 - Except for “fully meshed” conference: complexity just not worth it!
- ▶ **Terminology**
 - Trying to avoid already overloaded terms as much as possible
- ▶ **Functional entities in a “system model”**
 - Plus implementations examples
- ▶ **Set of protocols**
 - Definition of basic building blocks (SIP and other)
 - Sample combinations to implement conferencing services

Conference Participants

- ▶ **MUST work with basic SIP support only**
 - No awareness of conference
 - Just a point-to-point call with minimal means for control

- ▶ **Member types (SIP UA)**
 - **Basic participant:** plain-old SIP device
 - **Complex participant:** supports conferencing features
 - **Focus:** (one) center of a conference
 - **Anonymous:** Visible but unidentified participant
 - **Invisible:** Participant whose presence is not known

Conferences

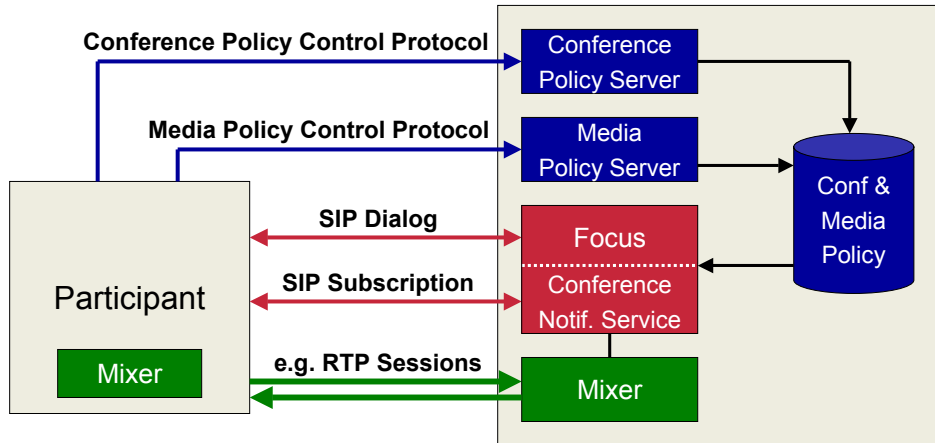
- ▶ **Conference types**
 - **Basic:** just plain SIP, no further means for control
 - **Complex:** some conferencing features provided
 - **Cascaded:** several foci concatenated in a conference
 - **Sidebar:** conference as (logical) part of another

- ▶ **Focus** Signaling center of a conference

- ▶ **Conference URI** Identifies a focus
(**isFocus** parameter may indicate this)

- ▶ **Factory URI** for automated conference creation
 - Yields a dynamically generated conference URI in return

Terminology and Model



Conferencing Scenarios (1)

- ▶ **Simple conferencing scenarios**
 - Plain SIP only (RFC 3261, 3264)
- ▶ **Extend point-to-point call**
 - Works only with local focus; otherwise, new call required
- ▶ **Ad-hoc conference**
 - Automated creation at focus
 - IVR / DTMF for control
 - Audio for information about the conference and its members
- ▶ **Reserved conference**
 - Same as ad-hoc
 - Use external means for reservation and configuration (e.g. web)

Conferencing Scenarios (2)

- ▶ **Advanced conferencing scenarios:**
 - Support for Call Transfer
 - Support means to communicate information from focus to UA
 - Optional: means to manipulate conference and media policy

- ▶ **Extend point-to-point call**
- ▶ **Join / create a conference based upon an existing dialog**
- ▶ **Ad-hoc conference**
- ▶ **Reserved conference**

- ▶ **Make use of additional conferencing features**

Sample Conferencing Features

- ▶ **Invite participants (dial-in, dial-out), expel participants**
- ▶ **Authenticate new participants by members**
- ▶ **Obtain conference and media policy information**
- ▶ **Manipulate **conference policy****
 - Participant privileges, participant management (black list, white list, ...)
 - Floor control,
- ▶ **Explicit media control (**media policy**)**
 - Configure media distribution
 - Add / remove media sessions
- ▶ **Create, control, and terminate sidebars**
 - Separate conference vs. media policy
- ▶ ...

Conference Policy

Define, retrieve/notify, modify, and act upon...

- ▶ **Formal rules for the conference**
 - Conference creation, termination, (policy) modification
 - Access control: black list, white list, rules for authentication
 - Privileges of individual participants
 - Visibility of the conference and its members
 - Access to floor and media policy (defined separately)
- ▶ **General conference attributes**
- ▶ **Participant management**
 - Invite, expel
- ▶ ...

Media Policy

- ▶ **Mixer model**
 - Input switch: collecting & selecting input streams from participants
 - Possibly transcoding and other per-media functions
 - Mixing topologies describing **mixing policies**
 - Output switch: selecting & distributing output streams to participants
 - Possibly transcoding and other per-media functions
- ▶ **Mixer may be centralized or not**
- ▶ **Media policy defines how incoming streams are processed, combined, and then distributed**
 - Individual mixing functions may be defined per participant
 - Common mixing functions may be defined for the conference
 - Mixing function may take into account “events” from other components

SIP Signaling Building Blocks

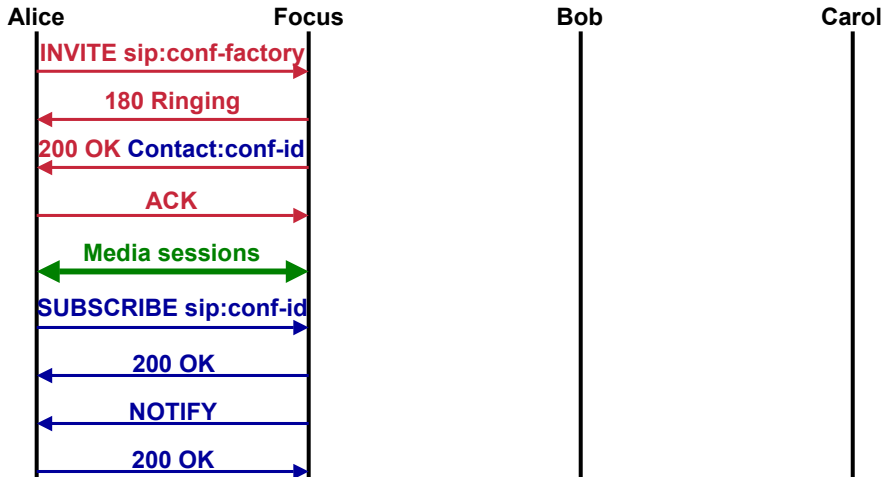
- ▶ **Membership control**
 - Initiation of conferences: INVITE
 - Inviting / adding to conferences: INVITE, REFER
 - Leaving a conference: BYE
 - Expelling from a conference: REFER (method="BYE")
- ▶ **Conference control**
 - State change notifications: SUBSCRIBE / NOTIFY
 - Dialog package, conference package, ...
 - Conference / media policy control
 - Might use data manipulation framework discussed in SIMPLE context
- ▶ **Other**
 - Determine focus URI OPTIONS
 - ...

Signaling Building Blocks using other Protocols

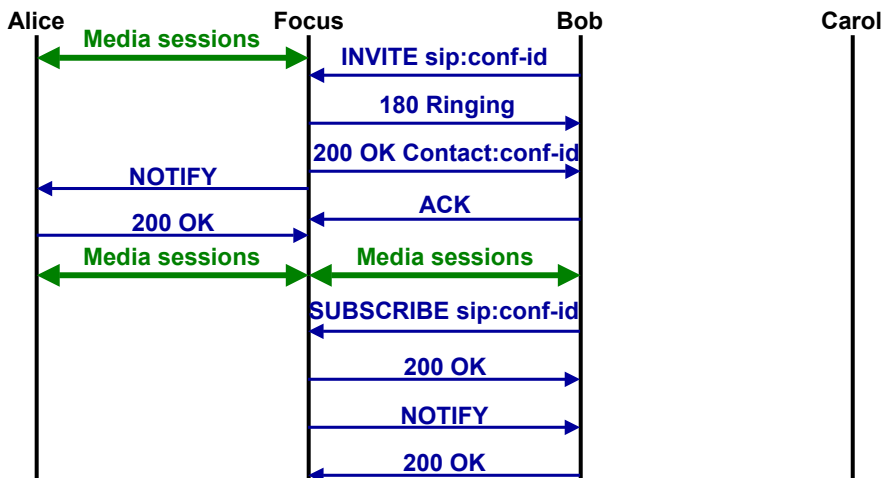
- ▶ **Conference Policy Control Protocol (CPCP)**
- ▶ **Media Policy Control Protocol (MPCP)**
- ▶ **Floor control protocol**

- ▶ **Common baseline:**
 - Request-response protocol (RPC-style)
 - May be strictly client – server
 - Asynchronous notifications provided by SIP
 - HTTP/HTML access to a web page for human interaction
 - SOAP RPCs e.g. over HTTP

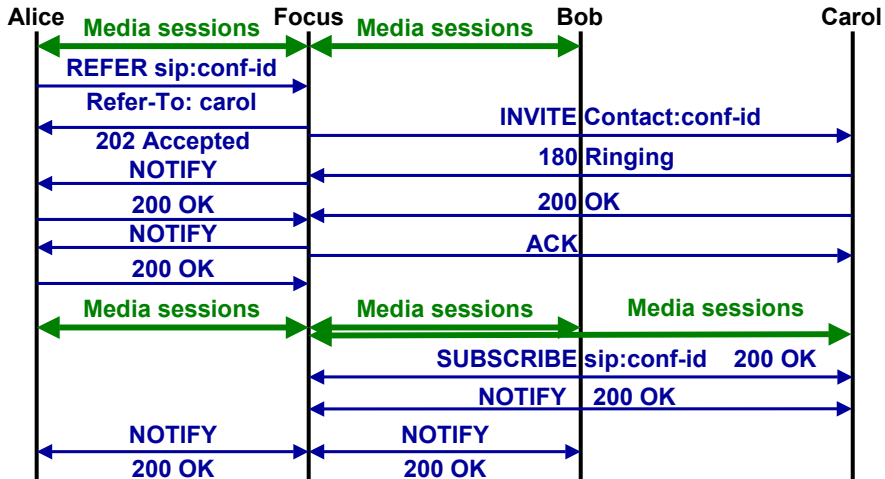
Call Flow Example: Conference Creation



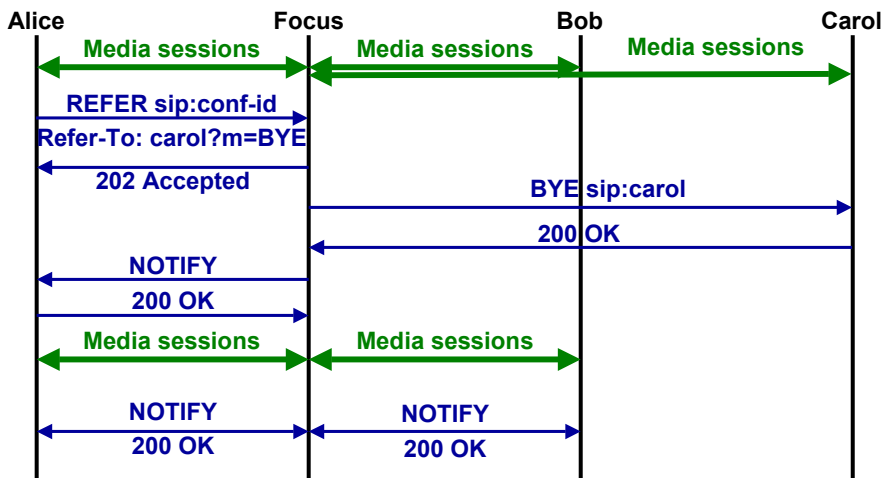
Call Flow Example: User Joining



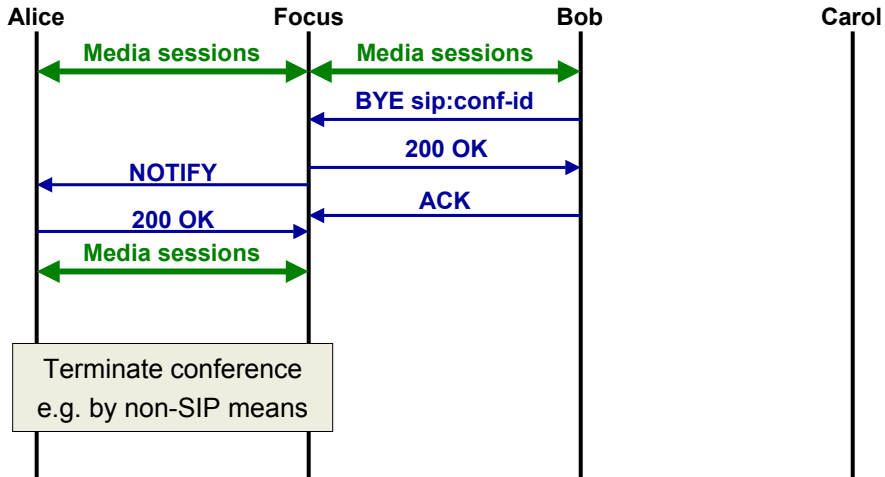
Call Flow Example: Adding a User



Call Flow Example: Removing a User



Call Flow Example: User Leaving



A Bit of Status

- ▶ **Framework: done**
- ▶ **Scenarios: agreed upon**
 - Issue: sidebar conferences
- ▶ **Requirements: nearing completion for many areas**
- ▶ **SIP Support: largely “complete”**
 - Still some minor issues
- ▶ **Conference and media control: moving ahead**
 - Requirements well advanced; protocol ideas progressing
- ▶ **Media control (e.g. “FIR”, Picture Freeze, ...)?**
- ▶ ...

Conclusion

- ▶ **After ~10 years of consideration, IETF conferencing seems to finally make it.**
- ▶ **Shift towards a more realistic target**
- ▶ **SIP as strong, commercially accepted base protocol**
- ▶ **Further control protocols developed at aggressive schedule**
- ▶ **Looking for further standardized applications**

- ▶ **Work in Progress**
- ▶ **Further info:**
<http://www.softarmor.com/sipping/teams/conf/>