



# Session Announcements

(SAP, RFC 2974)

# Session Description

(SDP, RFC 2327)

(SDP, RFC 4566)

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# Conference Establishment & Control

Session Description

## Workshop

### 1. Create

Descr.: IETF-Tag Internet-Multimedia  
 Orig.: J.Ott jo@tzi.org 327689113  
 Info: <http://www.tzi.org/dmn/>  
 Start: 29.09.2004 / 12:00  
 End: 29.09.2004 / 12:40  
 Media: Audio PCM 234.5.6.7/39000  
 Media: Video H.263 234.5.6.8/29000

### 2a. Announcement

- Announcement Protocol
- Netnews
- WWW

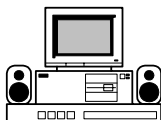
### 2b. Invitation

- E-Mail
- Invitation Protocol

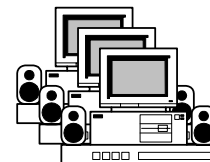
### 2c. Inquiry

- Streaming Protocol

↓ 3. Join

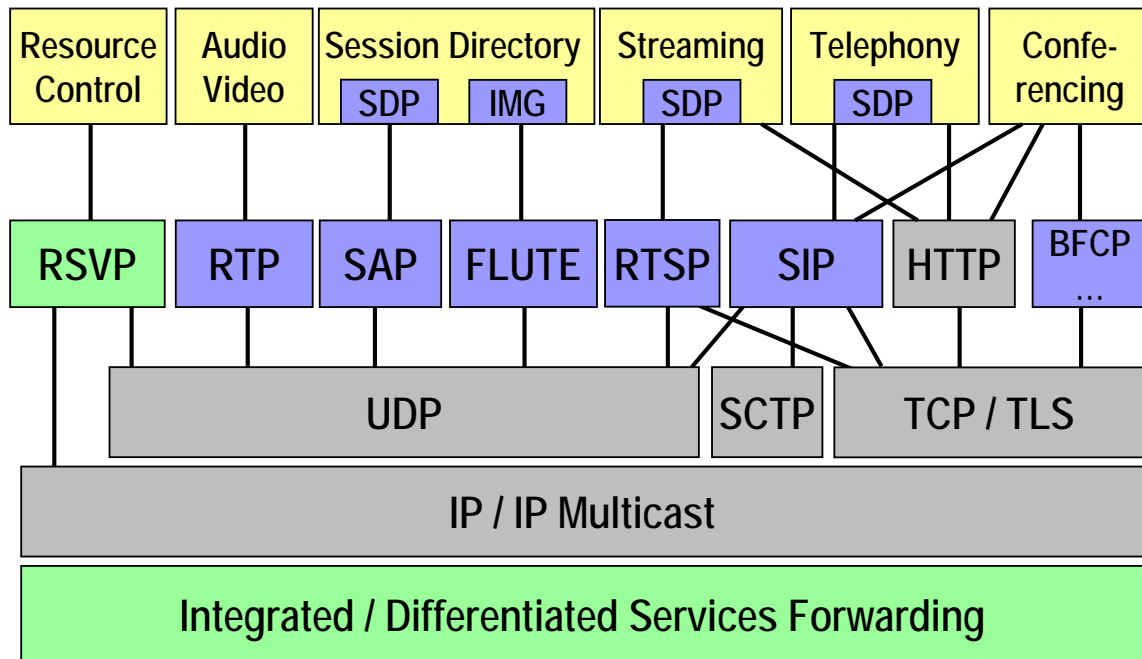


3. Join ↓ ↓ ↓





# IETF Multimedia (Conferencing) Architecture

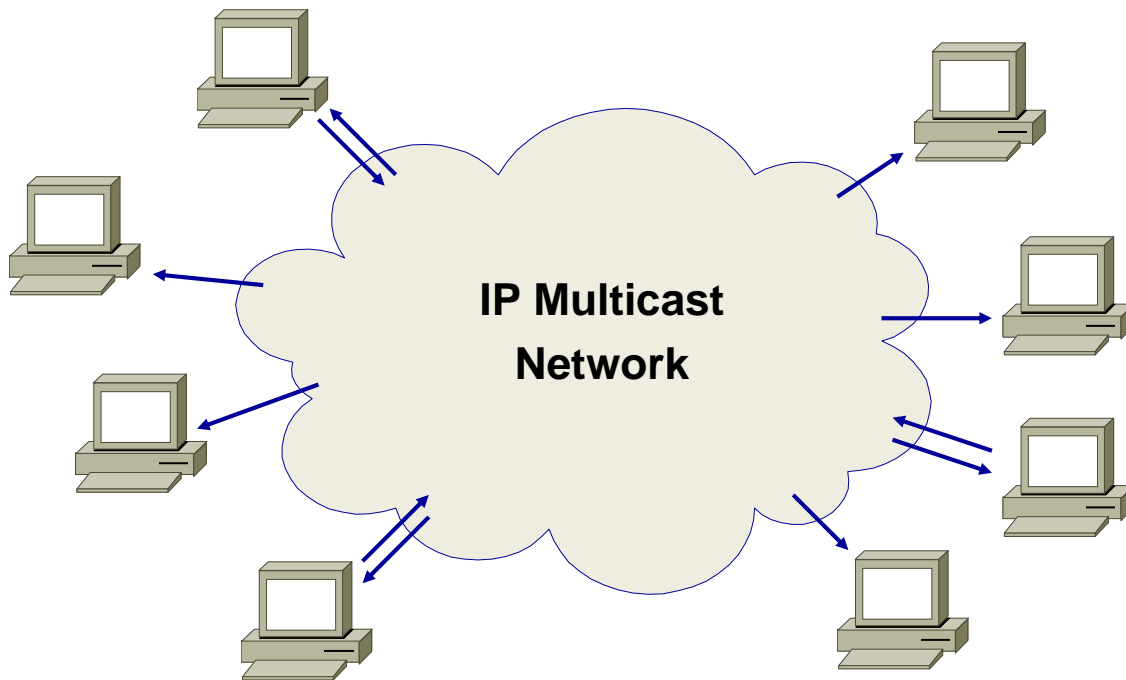


## Session Announcement Protocol (SAP)

- ▶ Announcing multimedia sessions to a broad audience
- ▶ Session announcements contain SDP
  - Subject of the session
  - Date(s) and time(s)
  - Media streams and addresses
  - Further information
- ▶ SAP Functions
  - New session announcements
  - Modify announcements
  - Delete announcements
  - Support for relays
- ▶ Earlier: Coordinate use of multicast address space



## SAP Scenario



## Dissemination of SAP Announcements

- ▶ **Scope of Announcements**
  - Per (administratively defined) multicast address scope
  - Local: 239.255.0.0/16
  - Organization local: 239.192.0.0/14
  - SAP conferences: 224.2.0.0 – 224.2.127.253
  - Other: Global
  - Similar considerations for IPv6
    - Scope identifier built-in into the IPv6 address structure
  
- ▶ **SDP descriptions should use addresses of same scope**
  - To ensure that receivers can also receive the media streams if they can receive the announcements



# SAP Features

- ▶ Limited announcement bandwidth per scope
  - e.g. 4000 bit/s (defined per scope)
  
- ▶ Calculation algorithm roughly similar to RTCP
  - Measure incoming SAP packets per scope
    - Sizes, number of announcements
  - Calculate size of own announcements
  - Estimate available share of bandwidth
  - Calculate own transmission interval
    - Use dithering ( $\pm 1/3$  of the interval)
    - Timer reconsideration before transmitting

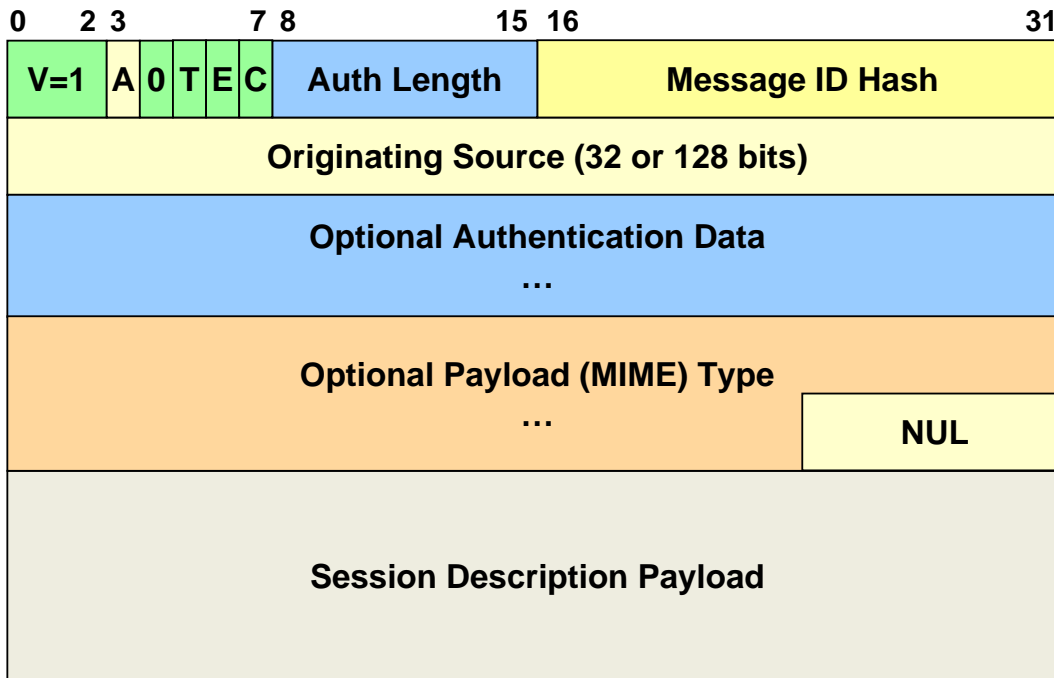


# (New) Announcements

- ▶ SAP uses UDP/IP: no reliability
- ▶ Repeat announcements in “regular” intervals
- ▶ Intervals: in the order of minutes
  - e.g. minimum 5 min
- ▶ Announcements for easy comparison identified by
  - Source IP address (of the creator)
  - 16 bit hash value
- ▶ May be authenticated (creator authentication)
- ▶ May be encrypted
- ▶ May be compressed
- ▶ May contain different payload types (SDP is just one)



# SAP Packet Format



## SAP Header Fields (1)

V: Version	—	=1	for SAPv2
A: Address type	—	=0	IPv4 source address
		=1	IPv6 source address
T: Type	—	=0	Announcement packet
		=1	Deletion packet
E: Encrypted	—		indicates encryption of the announcement packet
C: Compressed	—		indicates that the announcement packet is compressed
Auth Length	—		Length of the authentication header (0 = no authentication)



## SAP Header Fields (2)

Message ID Hash	—	Unique value per session creator
Originating Source	—	IP address of session creator
Authentication Data	—	Source Authentication information (PGP and CMS formats defined so far)
Payload MIME Type	—	NUL-terminated text string indicating the MIME type of the payload Default: application/sdp



## Deleting Announcements

- ▶ **Explicit Timeout**
  - No need to announce sessions after the “end time” in SDP
  - Caveat: the SAP receivers and relays need to understand SDP
- ▶ **Implicit Timeout**
  - Receiver observe repetition of announcement
  - After 10 times the announcement interval (or one hours) with re-announcement the session is removed
- ▶ **Explicit Deletion**
  - Send Deletion packet for a session
  - Message ID Hash and Originating Source must match
  - SHOULD be authenticated (match the original announcement)



## Modifying Announcements

- ▶ Replace an existing session description
  - E.g. modify media or start / end times
  - Update description
- ▶ Message ID Hash **MUST** change
- ▶ Modifying announcement **MUST** be authenticated if and only if the original announcement was
- ▶ If in doubt, a new session is “created”
  - Prevent denial-of-service attacks
- ▶ If proper match is found, the old session information is simply replaced by the new one

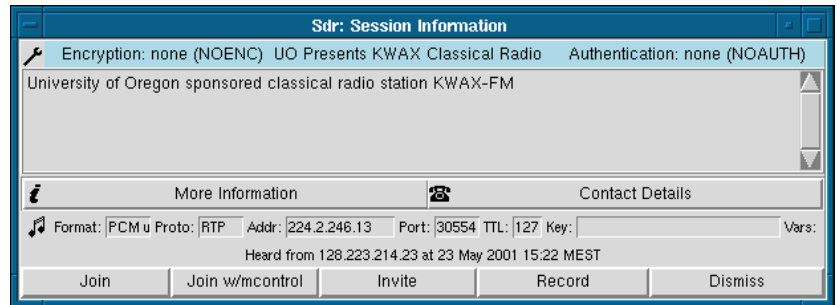
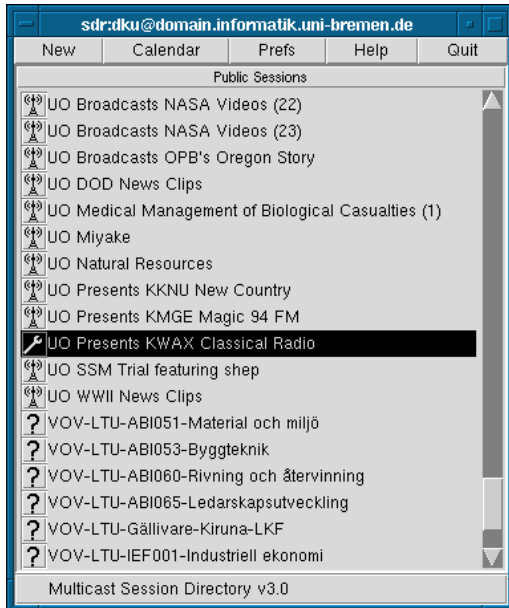


## SAP Security

- ▶ Encrypted messages for secure information distribution
  - Should be only used with limited size receiver groups
  - Avoid waste of computation resource if many receivers cannot decrypt the message
  - Key distribution out-of-scope
  - Limited applicability, limited usefulness
- ▶ Authentication
  - **SHOULD** always be done
  - Enables at least to verify that two messages are from the same source
  - Proper source authentication requires PKI
- ▶ General observation
  - Both is rarely used in practice
  - Current use of SAP in the Internet does not justify the effort...



# Session Announcement Tool: SDR



# Session Description Protocol (SDP)

- ▶ All you need to know about a session to join
  - who? — convener of the session + contact information
  - what about? — name and informal subject description
  - when? — date and time
  - where? — multicast addresses, port numbers
  - which media? — capability requirements
  - how much? — required bandwidth
- ▶ Grouped into three categories
  - 1 x session, m x time, n x media





## Session Level Description

- v=0 Version
- o= Owner / creator of the session + unique identifier + version
- u= URL for further information
- e= Contact email address
- p= Contact phone number
- b= Bitrate information
- k= Encryption key information
- z= Time zone adjustment
- a= Attribute lines (for extensions)
- c= Connection (=address) information



## Time Description

- ▶ Start and end time(s) of a session
  - Plus time zone adjustment
- ▶ Regular repetitions
  - Every Tuesday and Thursday, 10 – 12
  - Every day
- ▶ Arbitrary repetitions
  - Repeated specification of t= lines

t= Start, end time (NTP seconds, special case: 0, 0)

r= Repetitions (interval, duration, offsets)



# Media Description

## ▶ Define the media streams comprising a conference

- Media type (audio, video, text, tones, application, ...)
  - Only audio, video, text, tones are well-defined
- (multicast) address(es) + port number
- Maps RTP payload types for media to encoding formats
- Other media level attributes

m= Media and port specification

c= IP address specification (inherited from session)

a= Attributes for this media stream

rtpmap:, fmp:, recvonly, portrait | landscape



# SDP Example

## Length of Time represented by Media in a single Packet

(In SIP: address where originator wants to receive data)

```
v=0
o=llynch 3117798688 3117798739 IN IP4 128.223.214.23
s=UO Presents KWAX Classical Radio
i=University of Oregon sponsored classical radio station KWAX-FM
u=http://darkwing.uoregon.edu/~uocomm/
e=UO Multicastors multicast@lists.uoregon.edu
p=Lucy Lynch (University of Oregon) (541) 346-1774
t=0 0
a=tool:sdr v2.4a6
a=type:test
m=audio 30554 RTP/AVP 0
c=IN IP4 224.2.246.13/127
a=ptime:40
```

Session  
Level

Media  
Level



# Session Management Attributes

- ▶ Signaling the RTCP port (RFC 3605)
  - Motivation: RTP and RTCP port number may not be adjacent
  - `a=rtcp:<port> [<nettype> <addrtype> <addr>]`
  - `a=rtcp:60004 [IN IP4 192.168.11.12]`
- ▶ Signaling multicast sources (IMGPv3, SSM)
  - `a=src-filter:incl IN IP4 232.3.4.5 192.168.1.89`
  - `a=src-filter:excl IN IP4 225.3.4.5 192.168.1.89 192.168.6.66`
- ▶ Session bandwidth (independent of lower layers, RFC 3890)
  - `b=TIAS:64000`
  - `a=maxprate:40.0`
- ▶ RTCP bandwidth (modify sender/receiver share, RFC3556)
  - `b=RS:1600`
  - `b=RR:14400`



# Session Description and Capability Negotiation

From Session Announcement  
to Session Invitation



# Characteristics of SAP Announcements

## ▶ Common view

- Every SAP-receiver sees the same description
  - Session meta information & scheduling
  - Media description & transport parameters

## ▶ Identical transport parameters for all participants

- IP-Multicast service model:
  - Senders send to a multicast group (IP address)
  - Receivers join (“tune into”) a multicast group



# Session Initiation

## ▶ Distribute conference configuration

- Applications
  - Media types, media format parameters
- Transport Parameters
  - IP addresses, transport protocols, protocol parameters

## ▶ Negotiate Parameters!

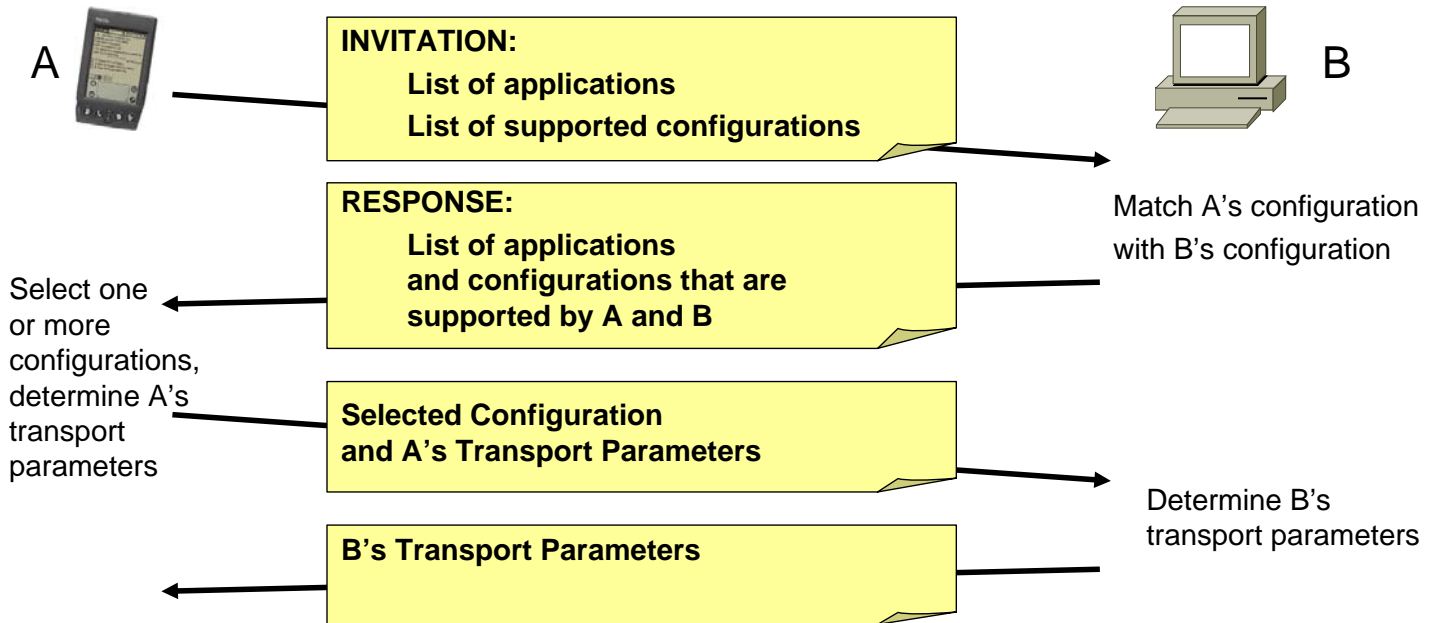
- Heterogeneous end systems
  - Different hardware and software capabilities
- User preferences

## ▶ SDP provides syntax mechanisms to express parameters

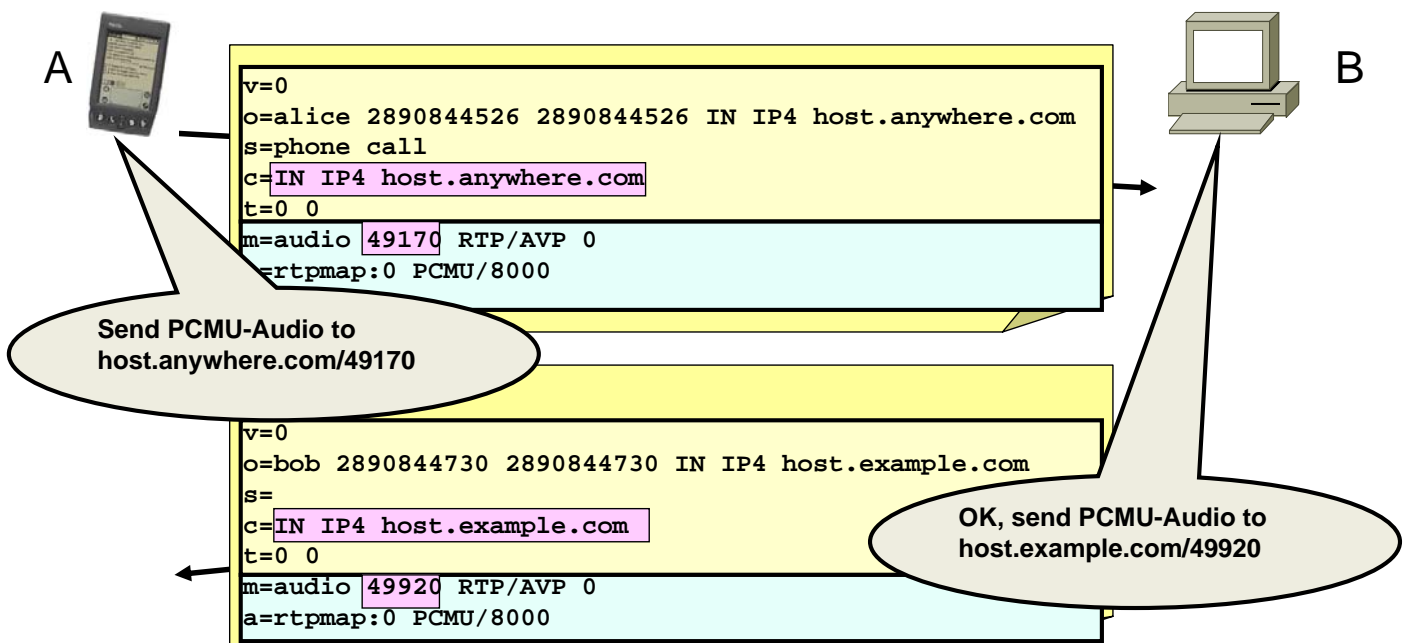
- Procedural model for initiation required



# Invitation: Conceptual Model

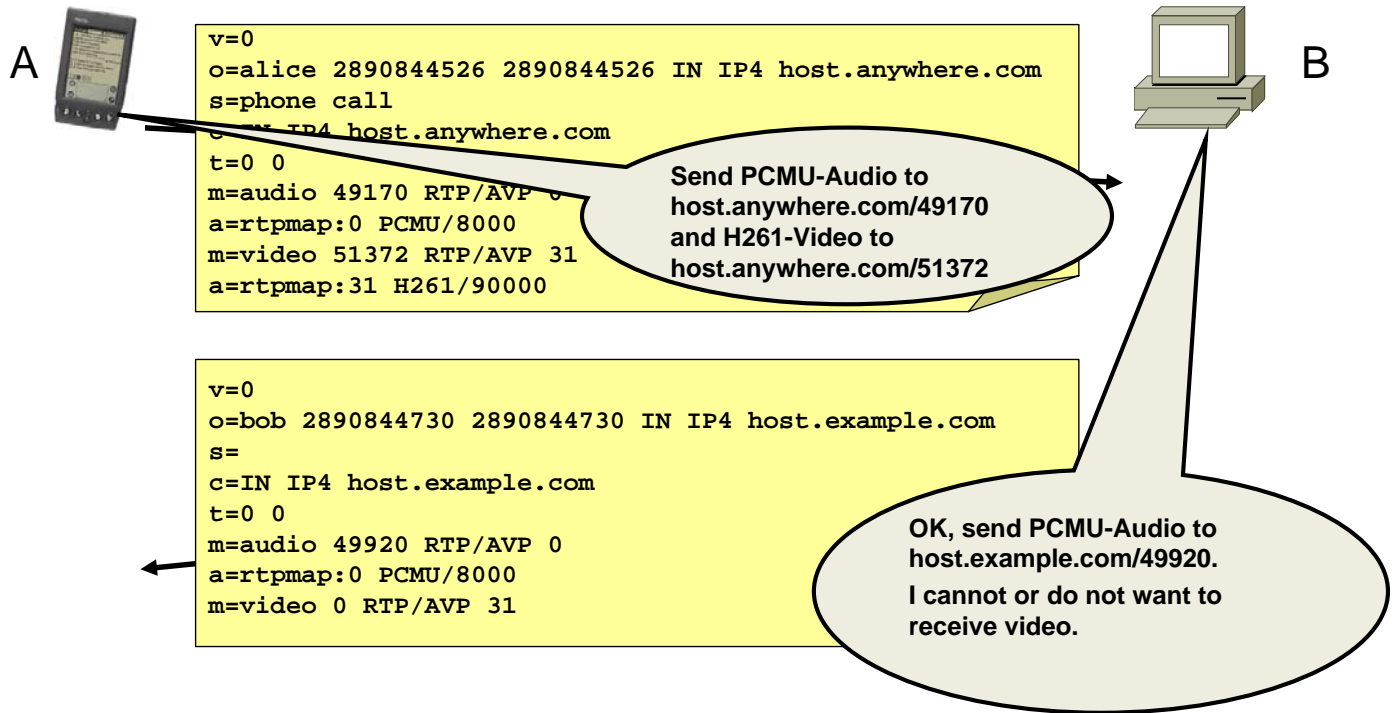


# Session Initiation with SDP (1)





## Session Initiation with SDP (2)

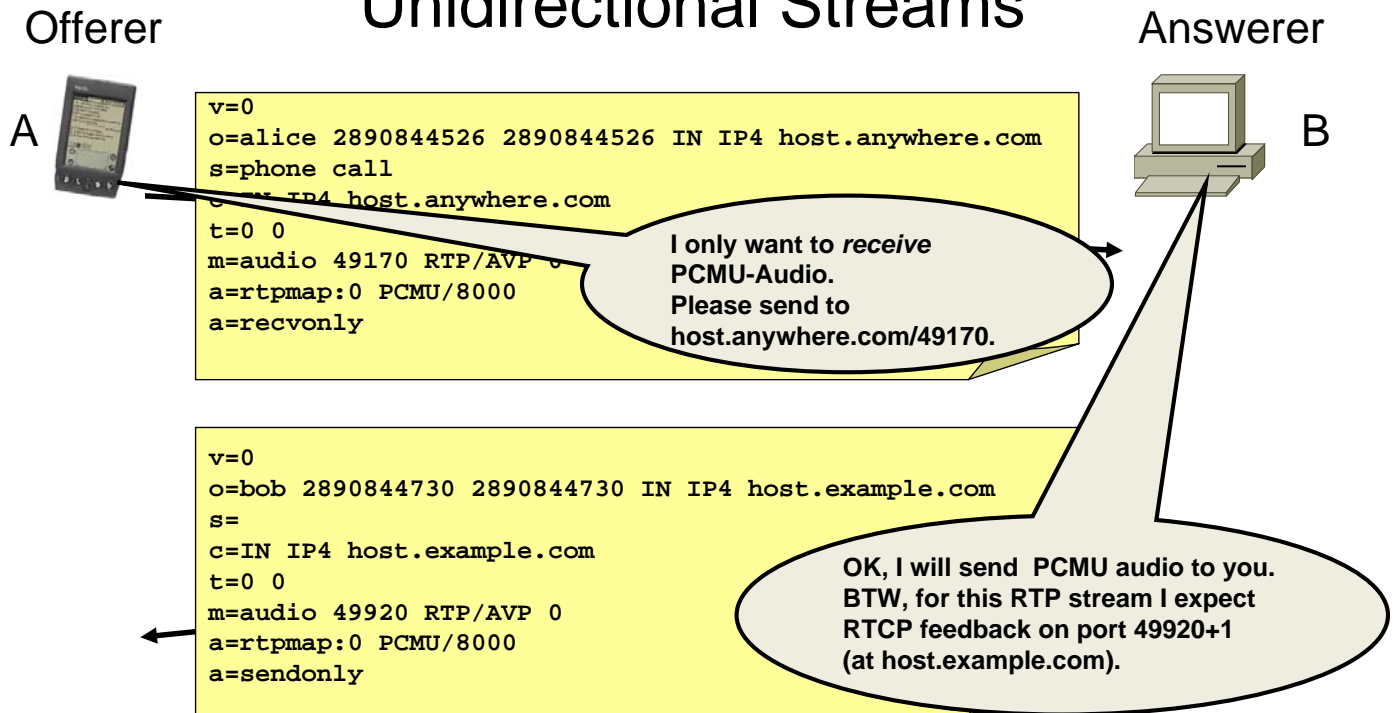


## SDP Offer/Answer Model (RFC 3264)

- ▶ For initiation of unicast sessions
- ▶ Objective: generate common view of session configuration
- ▶ Simple exchange of capability descriptions
  
- ▶ Basic Model:
  - A sends offer to B, including
    - Set of media streams and codecs A wishes to use
    - Transport parameters (where A wants to *receive* data)
  - B sends answer to A
    - For each stream in offer, indicating whether stream is accepted or not
    - For each stream add transport parameters (where B wants to *receive* data)



# Unidirectional Streams



# Send/Receive Only

- ▶ Media streams may be unidirectional
  - Indicated by *a=sendonly*, *a=recvonly*
- ▶ Attributes are interpreted from sender's view
- ▶ *sendonly*
  - Recipient of SDP description should not send data
  - Connection address indicates where to send RTCP receiver reports
  - Multicast session: recipient sends to specified address
- ▶ *recvonly*
  - Sender lists supported codecs
  - Receiver chooses the subset he intends to use
  - Multicast session: recipient listens on specified address
- ▶ *inactive*
  - To pause a media stream (rather than deleting it)

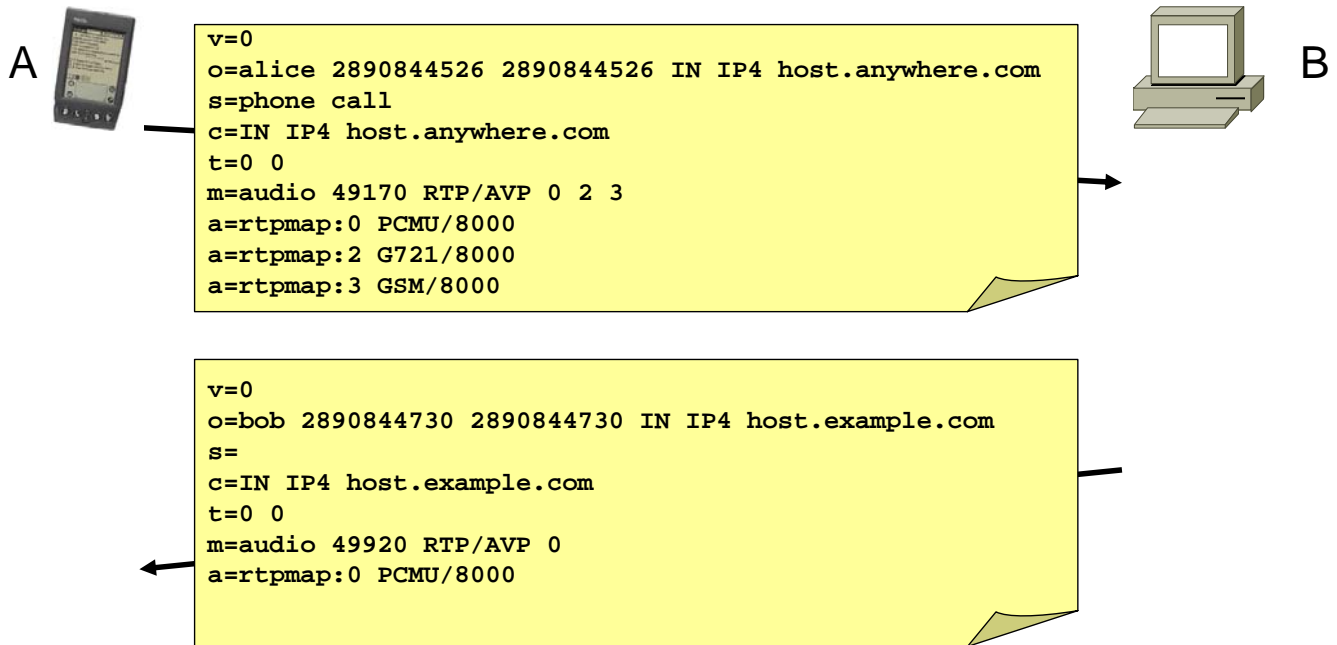




# Codec Selection

Offerer

Answerer

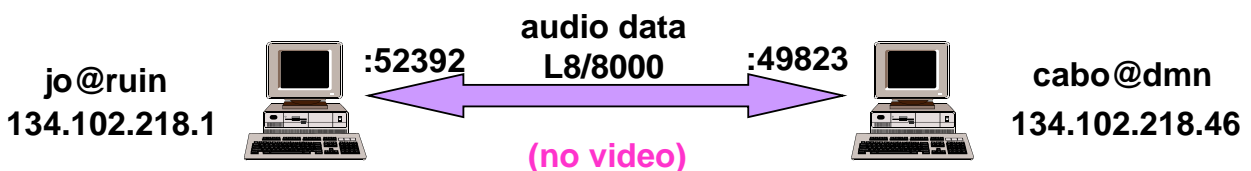


# Example SDP Alignment

```
v=0
o=jo 7849 2873246 IN IP4 ruin.inf...
s=SIP call
t=0 0
c=IN IP4 134.102.218.1
m=audio 52392 RTP/AVP 98 99
a=rtpmap:98 L8/8000
a=rtpmap:99 L16/8000
m=video 59485 RTP/AVP 31
a=rtpmap:31 H261/90000
```

```
v=0
o=cabo 82347 283498 IN IP4 dmn.inf...
s=SIP call
t=0 0
c=IN IP4 134.102.218.46
m=audio 49823 RTP/AVP 98
a=rtpmap:98 L8/8000
m=video 0 RTP/AVP 31
```

Resulting configuration:





## Grouping of m= lines in SDP

### ▶ Observation:

- Multiple m= lines in SDP have no relationship to each other
  - Independent media streams
  - usually different media types

### ▶ Problem:

- Want to express synchronization relationship
  - Lip synchronization
- Concept of “flows” that consist of several media streams
  - Streams encoded in several formats
  - May be streamed from different hosts/ports
  - Useful application in some IP telephony scenarios



## Example for Lip Synchronization

Stream 1 and 2  
should be  
synchronized.

```
v=0
o=Laura 289083124 289083124 IN IP4 one.example.com
t=0 0
c=IN IP4 224.2.17.12/127
a=group:LS 1 2
m=audio 30000 RTP/AVP 0
a=mid:1
m=video 30002 RTP/AVP 31
a=mid:2
m=audio 30004 RTP/AVP 0
i=This media stream contains the Spanish translation
a=mid:3
```



# ANAT Grouping

- ▶ **Alternative Network Address Types (RFC 4091)**
  - Allows expressing IPv4 and IPv6 address alternatives

```
v=0
o=bob 280744730 28977631 IN IP4 host.example.com
s=
t=0 0
a=group:ANAT 1 2
m=audio 25000 RTP/AVP 0
c=IN IP6 2001:DB8::1
a=mid:1
m=audio 22334 RTP/AVP 0
c=IN IP4 192.0.2.1
a=mid:2
```



# FEC Grouping

- ▶ **Group basic and FEC data (draft-ietf-mmusic-fec-grouping-05.txt)**

```
v=0
o=adam 289083124 289083124 IN IP4 host.example.com
s=ULP FEC Seminar
t=0 0
c=IN IP4 224.2.17.12/127
a=group:FEC 1 2
a=group:FEC 3 4
m=audio 30000 RTP/AVP 0
a=mid:1
m=application 30002 RTP/AVP 100
a=rtpmap:100 ulpfec/8000
a=mid:2
m=video 30004 RTP/AVP 31
a=mid:3
m=application 30004 RTP/AVP 101
c=IN IP4 224.2.17.13/127
a=rtpmap:101 ulpfec/8000
a=mid:4
```



## Further Groupings

- ▶ Alternative RTP profiles
  - Dealing with combinatorial explosion of options
  - E.g. AVP and AVPF, AVP and SAVP
- ▶ Layered coding and scalable (video) coding
  - Convey dependencies across different RTP sessions
- ▶ Alternative addresses?
  - Multicast vs. unicast distribution, e.g. for a media streaming offer
- ▶ ...
- ▶ Currently design team formed
  - Resulting solution may but need not use grouping



## Simple Capability Declaration in SDP

- ▶ Observation:
  - Capability negotiation/declaration in SDP too limited
  - Session description describe both session parameters and capabilities without clear distinction
  - Simultaneous capability restrictions cannot be expressed
    - *“Supporting multiple codecs for one media type, but only one per session”*
- ▶ Simcap: add SDP attributes to explicitly express capabilities



# Simcap Example

Sender is willing  
to receive and send  
G.729 (18)  
and telephone-events.

Additionally, it declares the  
following capabilities:

- PCMU-Audio (0)
- telephone-events (different events)
- Fax-Relay over UDP and TCP

```
v=0
o=- 25678 753849 IN IP4 128.96.41.1
s=
c=IN IP4 128.96.41.1
t=0 0
m=audio 3456 RTP/AVP 18 96
a=rtpmap:96 telephone-event
a=fmtp:96 0-15,32-35
a=sqn: 0
a=cdsc: 1 audio RTP/AVP 0 18 96
a=cpar: a=fmtp:96 0-16,32-35
a=cdsc: 4 image udptl t38
a=cdsc: 5 image tcp t38
```



# Simcap Example

Semantics:

- a=sqn: declares a sequence number
- a=cdsc: declare one or more capabilities
- a=cpar: additional parameters for a declaration

```
v=0
o=- 25678 753849 IN IP4 128.96.41.1
s=
c=IN IP4 128.96.41.1
t=0 0
m=audio 3456 RTP/AVP 18 96
a=rtpmap:96 telephone-event
a=fmtp:96 0-15,32-35
a=sqn: 0
a=cdsc: 1 audio RTP/AVP 0 18 96
a=cpar: a=fmtp:96 0-16,32-35
a=cdsc: 4 image udptl t38
a=cdsc: 5 image tcp t38
```



# Connection-oriented Media with SDP

- ▶ Focus on TCP (RFC 4145) and TLS (RFC 4572)
- ▶ In contrast to UDP, a connection must be established
  - Who is to initiate setup, who is to listen?
    - `a=setup: active | passive | actpass | holdconn`
  - What if a connection already exists (e.g., when renegotiating)
    - Keep the existing connection?
    - Set up a new one?
    - `a=connection: new | existing`
  - When to tear down a connection?
    - If a “new” one is specified, close an existing one
- ▶ Relies on interactive agreement on how to proceed



# Labeling media streams

- ▶ Unique identification
  - Across SDP session descriptions
    - Contrast to mid (which is valid within a session only)
  - `a=label:<token>`
  - No semantics
- ▶ Attaching stream semantics
  - Usually relevant within an SDP session
  - Hint at stream semantics
    - E.g., if multiple media streams are received: which is which?
  - `a=content:<token>`
  - `token=slides | speaker | s1 | main | alt | user-floor | ...`



## SDP Extensions: There is more...

- ▶ Precondition signaling for media streams
  - Security
  - QoS
  - Connectivity
- ▶ Key management (fixing k=)
  - End-to-end key negotiation
  - End-to-end key distribution (via a protected channel)
- ▶ And support for further media types
  - Multicast file distribution, application sharing, ...
- ▶ Will be discussed in the context of signaling protocols



## Summary So Far

- ▶ SDP syntax can be used for session initiation
  - But requires additional specification of procedures: Offer/Answer
- ▶ SDP & Offer/Answer not appropriate for all usage scenarios
  - Fundamental SDP problem of combining configuration descriptions with capability declaration
  - Lack of expressiveness: grouping of media streams
  - "a=" only a limited extension mechanism
- ▶ SDP Syntax
  - Limited expressiveness and cumbersome extensibility



# SDP Syntax Issues

- ▶ **Basic** set of description elements for media sessions
  - IP addresses, port numbers, RTP payload types, parameters
- ▶ **Extensibility**: new session / media level attributes
  - `a=<keyword>:<value> ...`
  - Senders can use arbitrary attributes:
    - Important attributes cannot be distinguished from unimportant ones
    - Name clashes (misinterpretation) cannot be excluded
  - In principle, allows for any kind of extension
    - Grouping, constraints, ...
- ▶ SDP workarounds rather clumsy, inefficient, ...



# SDP Next Generation (SDPng)

- ▶ Being designed to address SDP's flaws...
  - Limited expressiveness
    - For individual media and their combination
    - Often only very basic media descriptions available
  - No real negotiation functionality
  - Limited extensibility (clumsy, hard to coordinate)
  - No semantics for media sessions (only implicit)
  - Also: Avoid second system syndrome!
    - Simple, easy to parse, extensible, limited scope
- ▶ Major issue: syntax choice (XML)
  - Not backwards-compatible (deployment, vendor know-how, code re-use)
  - Back in the late 1990s, XML considered "too expensive" for endpoints
- ▶ New approaches to capability negotiation presently discussed





# Intelligent Endpoints



- ▶ Intelligent endpoints with support for
  - Multiple codecs and format parameters
  - Different applications (e.g., audio, DTMF, video, games)
  - Many transport parameters
    - RTP/UDP/IPv4, RTP/UDP/IPv6, Security, Source-Specific-Multicast...
  - AAA & security parameters

Must be expressible in configuration descriptions!



# Intelligent Endpoints



- ▶ Heterogeneous end systems
  - Different capabilities
  - Different user preferences
  - Dynamic configuration

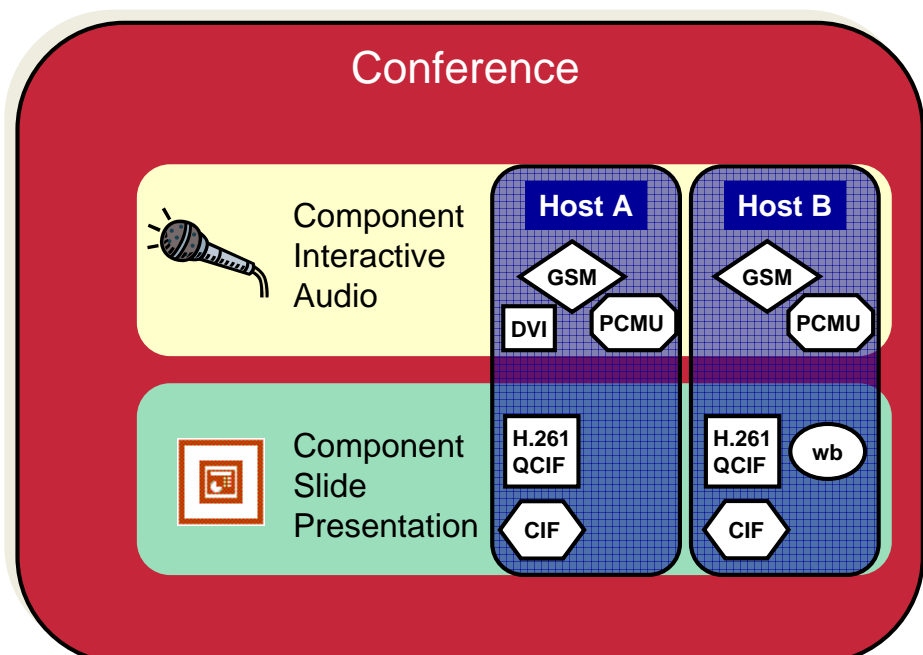
Interoperability requires dynamic negotiations of parameters!

# Specific Requirements

- ▶ Expressiveness
  - Describe all **required** configuration parameters
- ▶ Extensibility
  - No fixed parameter set
  - Profiles (“packages”) for new configuration parameters
- ▶ Support for Negotiation
  - Derive commonly supported configurations from individual configuration descriptions (for  $n \geq 2$ )
- ▶ Compatibility
  - Drop-in replacement for SDP in SIP applications

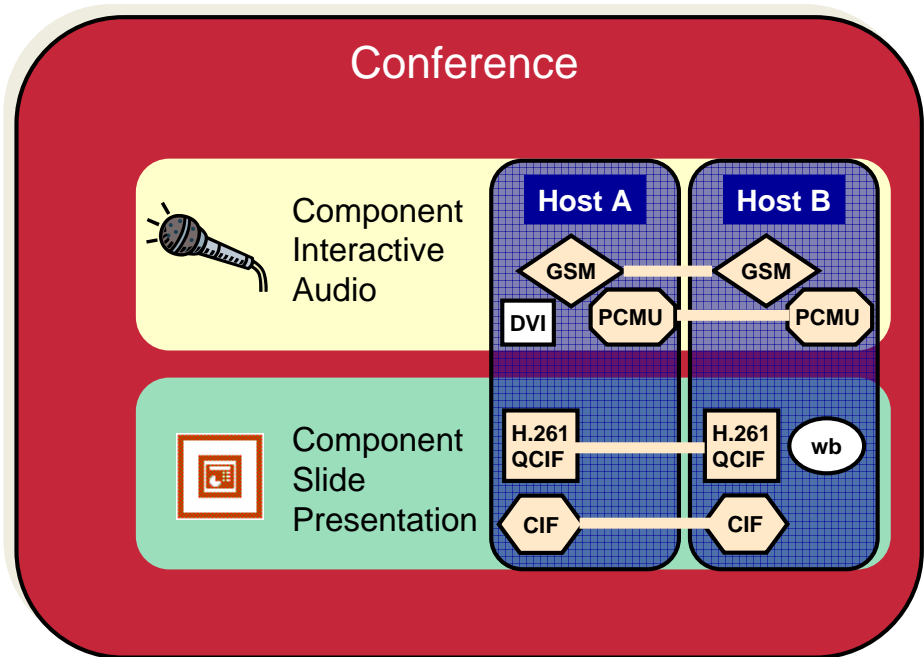
# SDPng's Conference Model

- ▶ Components in a conference
  - Individual cooperation functions
  - Characterized by the service they provide (not by their technical implementation)
- ▶ Implementations of components
  - Depend on endpoint capabilities and user preferences
  - Use of implementations must be configured or negotiated



# Potential Configurations

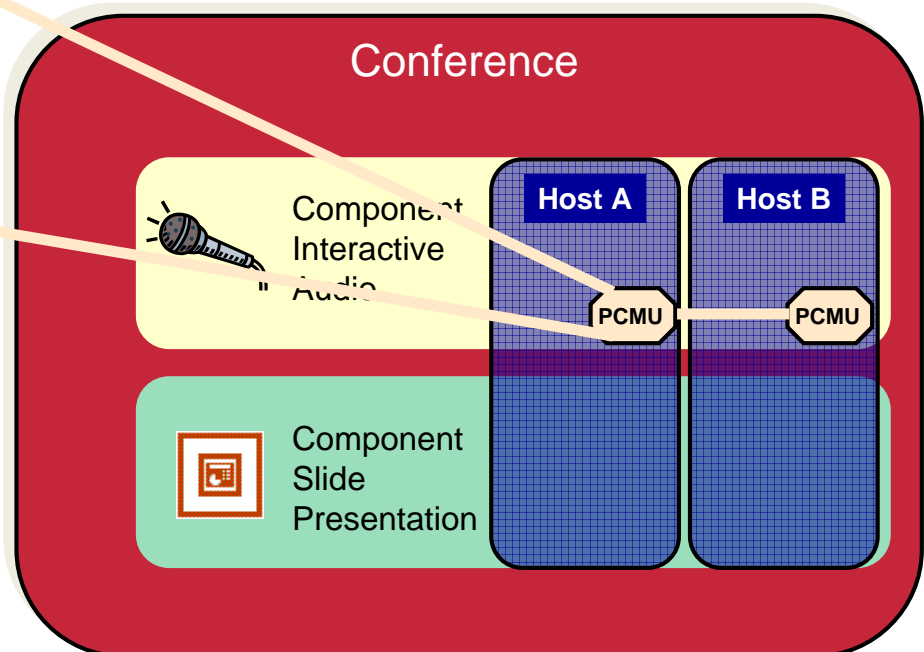
- ▶ Configurations for implementing a component
  - Common capabilities
  - Not a complete conference description, e.g., no transport parameters
  - Dynamic set of parameters
    - Can change over the course of a conference



# Actual Configurations

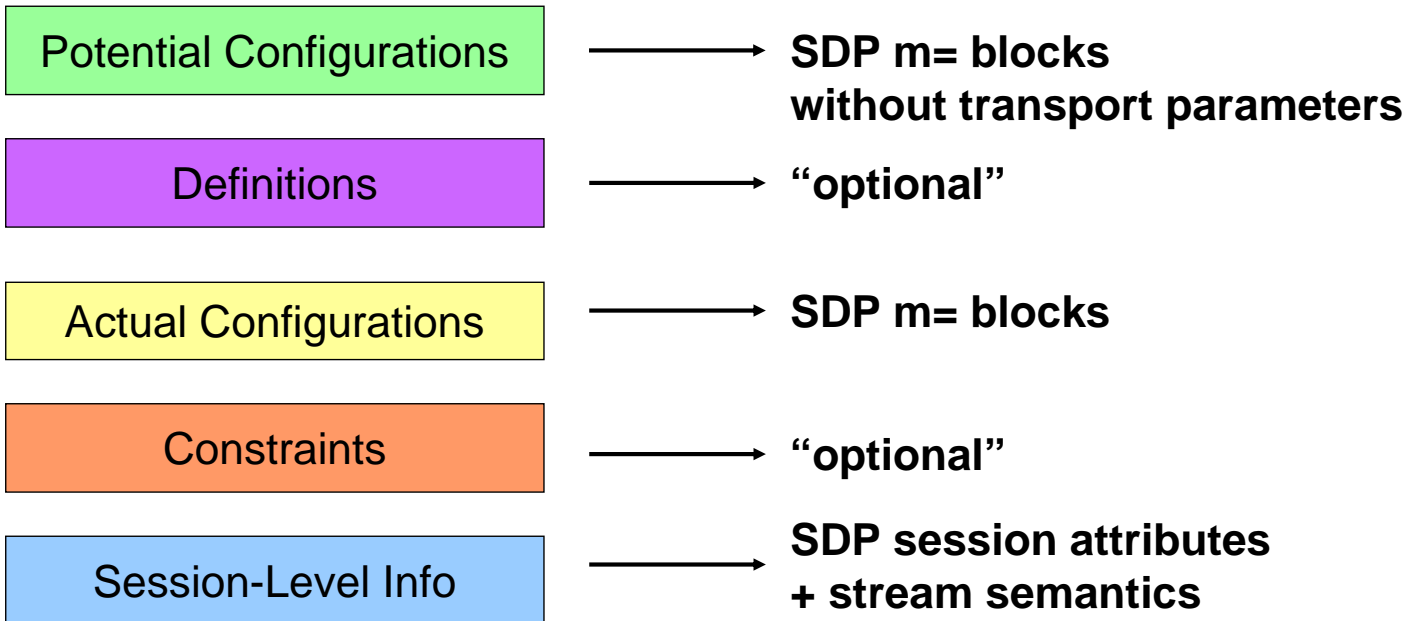
```
address=192.168.1.1
port=37000
codec-type=PCMU
payload-type=0
...
```

- ▶ Complete specification of conference parameters
  - Selected subset of potential configurations
  - Complemented with
    - Media format parameters
    - Transport parameters

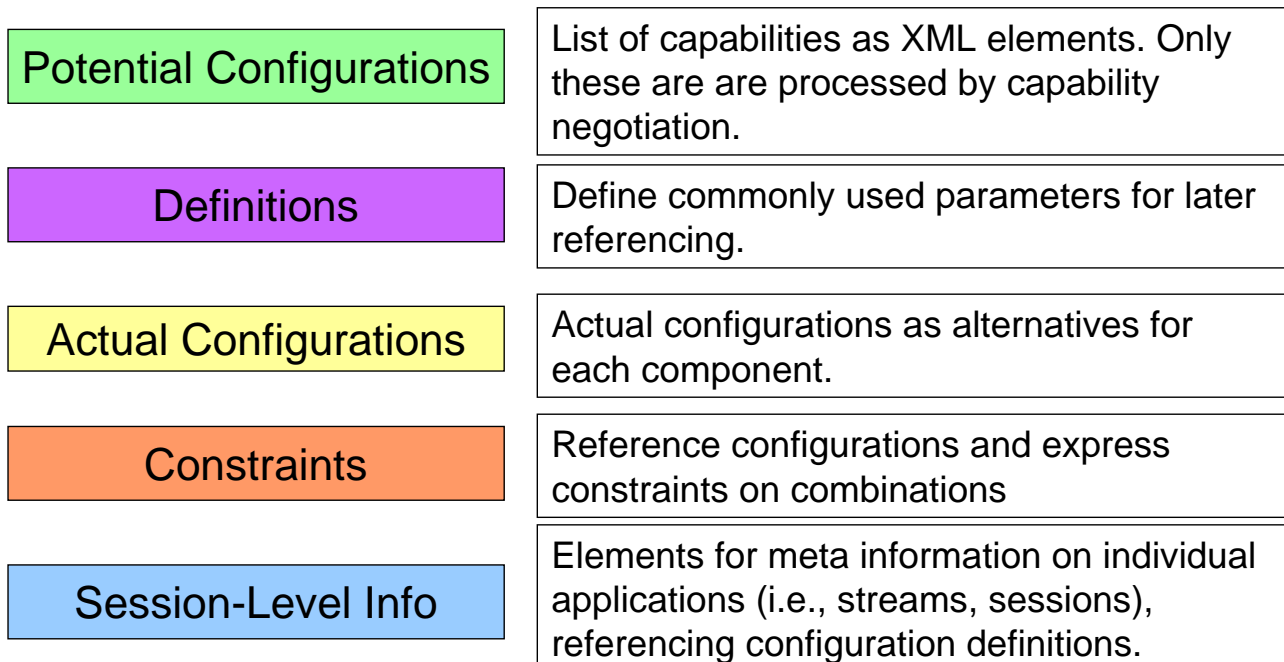




## General SDPng Model



## SDPng Structure





# SPDng: An Extensible Framework

## SPDng consists of

- Base specification
    - Overall structure of SPDng documents
    - Common data types and element types
  - Basic rules packages (“profiles”)
    - Define how to express commonly used parameters
      - Codecs, RTP parameters etc.
- Formally specified**
- Basic definitions (“libraries”)
    - Specific codec definitions, RTP payload type definitions etc.
- SPDng description instances**



# Capability Model

- ▶ Three different types
  - Tokens:
    - `encoding=PCMU`
    - `Ascertain identity || fail`
  - Token lists:
    - `sampling-rate=8000,16000, 44000`
    - `Determine common subset || fail`
  - Numerical Ranges
    - `6 <= bitrate <= 64`
    - `Determine common sub-range || fail`
- ▶ Distinguish *optional* capabilities
  - `silence-suppression supported`
  - Applicable to each type, failing results in removing the capability, interoperability still possible



# XML Syntax (1)

- ▶ Feature independent negotiation
  - Process capability descriptions without knowing semantics
  - Access to schema definition not required



# XML Syntax (2)

- ▶ Capabilities
  - A collection of independent definitions
  - Each definition is processed independently
  - Every property is a single XML element
    - Tokens and token lists as element content
    - Numerical ranges with explicit XML attributes
    - No further substructure
    - Descriptions are still standalone

```
<audio:codec name="avp:pcmu">  
  <audio:encoding>PCMU</audio:encoding>  
  <audio:channels>1 2</audio:channels>  
  <audio:sampling>8000 16000</audio:sampling>  
  <audio:bitrate min="6" max="64"/>  
  <audio:silence-suppression status="opt"/>  
</audio:codec>
```



# Formal Schema Definition

## ▶ Base specification

- SDPng XML document structure
- Basic data types (token, token lists, ranges)
- XML-Schema as a definition mechanism

## ▶ Package definitions

- Application specific vocabulary
- Each package definition in unique XML namespace
- XML-Schema as a definition mechanism



```
<xsd:complexType name="audio:CodecT">
  <xsd:complexContent>
    <xsd:extension base="sdpng:Definition">
      <xsd:sequence>
        <xsd:element name="encoding" type="sdpng:token"/>
        <xsd:element minOccurs="0" name="channels"
          type="sdpng:tokenlist"/>
        <xsd:element minOccurs="0" name="sampling"
          type="sdpng:tokenlist"/>
        <xsd:element minOccurs="0" name="bitrate"
          type="sdpng:range"/>
        <xsd:element minOccurs="0" name="silenceSuppression"
          type="sdpng:optToken"/>
      </xsd:sequence>
    </xsd:extension>
  </xsd:complexContent>
</xsd:complexType>

<xsd:element name="audio:codec" type="audio:CodecT"
  substitutionGroup="sdpng:definition"/>
```

## Sample Package Definition



## Specifying Configurations (1)

```
<cap>
  <audio:codec name="avp:pcmu">
    <audio:encoding>PCMU</audio:encoding>
    <audio:channels>1 2</audio:channels>
    <audio:sampling>8000 16000</audio:sampling>
    <audio:bitrate min="6" max="64"/>
    <audio:silence-suppression status="opt"/>
  </audio:codec>
  <rtp:udp name="rtpudpip6">
    <rtp:network>IP6</rtp:network>
  </rtp:udp>
</cap>
```



## Specifying Configurations (2)

```
<cap>
  <audio:codec name="avp:pcmu"> [...]</audio:codec>
  <rtp:udp name="rtpudpip6"> [...] </rtp:udp>
</cap>

<def>
  <rtp:udp name="rtp-cfg1" ref="rtp:rtpudpip6">
    <rtp:ip-addr>::1</rtp:ip-addr>
    <rtp:port>9456</rtp:port>
    <rtp:pt>1</rtp:pt>
  </rtp:udp>
</def>
```





## Specifying Configurations (3)

```
<cap>
  <audio:codec name="avp:pcmu"> [...] </audio:codec>
  <rtp:udp name="rtpudp6"> [...] </rtp:udp>
</cap>
<def>
  <rtp:udp name="rtp-cfg1">[...]</rtp:udp>
</def>

<cfg>
  <component name="interactive-audio" media="audio">
    <alt name="alt1">
      <audio:codec ref="avp:pcmu"/>
      <rtp:udp ref="rtp-cfg1"/>
    </alt>
  </component>
</cfg>
```



## Specifying Configurations (4)

- ▶ Each *component* (application session) element provides list of alternatives
- ▶ Each *alternative* provides definitions for the *component*
  - Referencing definitions from the capability section
    - Providing additional parameters, where required
    - Alternatives that reference non-interoperable definitions are discarded
  - List of definitions
    - No nesting of elements from different packages
  - Semantics are application-specific
    - Applications MUST know how to interpret definitions
  - No restrictions on quantity or order



# Libraries

- ▶ Libraries:
  - Pre-defined definitions, e.g., a set of audio codec definitions
  - Referenced from a description document
  
- ▶ Semantics difficult to get right
  - Application-independent negotiation would require access to library definitions
    - Requirement to *include* library definitions into description document
    - Capability negotiation has to consider *all* definitions
  
- ➔ Forego libraries, include definitions inline



# Summary

- ▶ Extensibility and dynamic negotiation are key to interoperability
  - Intelligent endpoints and new services require a capable and flexible description mechanism
  
- ▶ SDPng to provide interoperability *and* extensibility
  - Simple applications stay simple
  - Innovation is possible through structured extensibility
  
- ▶ Smooth migration from SDP to SDPng is possible
  - “Bi-lingual” endpoints and mapping of SDP to SDPng