



Announcements

- ▶ NO LECTURE next Tuesday (4.12.)
- ▶ NO LECTURE on Thursday (13.12.)
- ▶ Assignment 3: Next Wednesday (5.12.)
- ▶ Read
 - A Framework for Application Interaction in SIP
 - draft-ietf-sipping-app-interaction-framework-05.txt
 - A SIP Event Package for Key Press Stimulus (KPML)
 - RFC 4730
 - SIP for telephony slides
- ▶ Updated schedule
 - 11.12. Jabber, SIP for telephony, and conferencing
 - 12.12. SIP in the real-world, exam hints



SIP Extensions and Service Creation



Basic Extension Mechanisms

- ▶ Proxies forward unknown methods and headers
 - ▶ UAS' ignore unknown headers, reject methods

 - ▶ Feature negotiation
 - Headers: **Require**, **Proxy-Require**, **Supported**
 - Option tags for feature naming (see below)
 - Error responses:
 - 405 Method not allowed
 - 420 Unsupported
 - 421 Extension Required
- } **UAC and UAS/proxy must agree on common feature subset**
-
- ▶ Option tags
 - Identified by unique token
 - Prefix reverse domain name of creator
 - IANA: implicit prefix *org.ietf*.



Some SIP Extensions

- ▶ Reliable provisional responses
- ▶ Session Timers
- ▶ Early Media
- ▶ Adjusting session state: UPDATE
- ▶ INFO method
- ▶ REFERing peers to third parties
- ▶ SIP for subscriptions and event notifications
- ▶ Instant messaging
- ▶ SIP for presence
- ▶ ...

- ▶ Addressed later today where appropriate...



Reliable Provisional Responses

- ▶ Signaling Gateways, ACD systems etc. may depend on provisional responses
 - Such as 180 Ringing

→ need optional reliability

- ▶ Option tag 100rel:
 - End-to-end reliability of provisional responses > 100
 - Retain order of reliable responses
- ▶ Implementation:
 - Windowing: RSeq/RAck headers
 - Explicit acknowledge: PRACK request

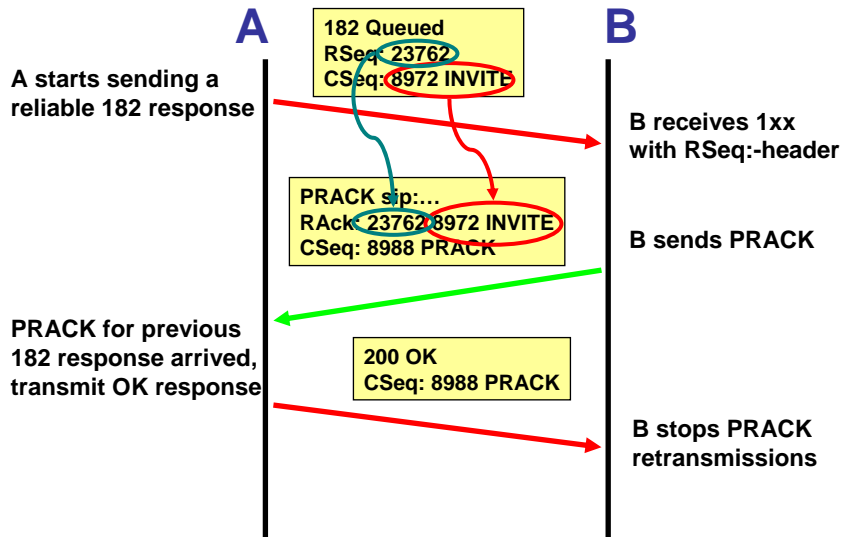


Use of PRACK

- ▶ Insert RSeq:-header with random integer value
 - new responses must increment RSeq: by 1
- ▶ Retransmit with exponential backoff
- ▶ No subsequent reliable provisional responses until first PRACK
 - enables re-ordering at receiver side
- ▶ PRACK request
 - RACK: contains RSeq: and CSeq: values to acknowledge
 - May contain body with additional information



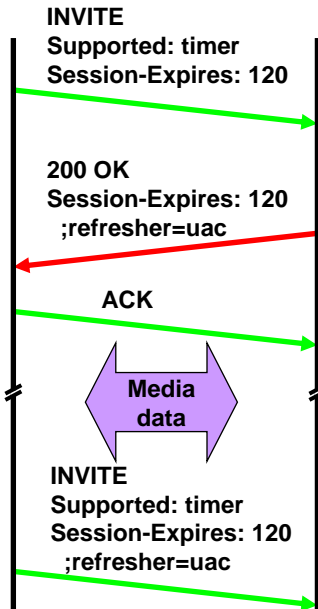
Example Call Flow: PRACK



Session Timers

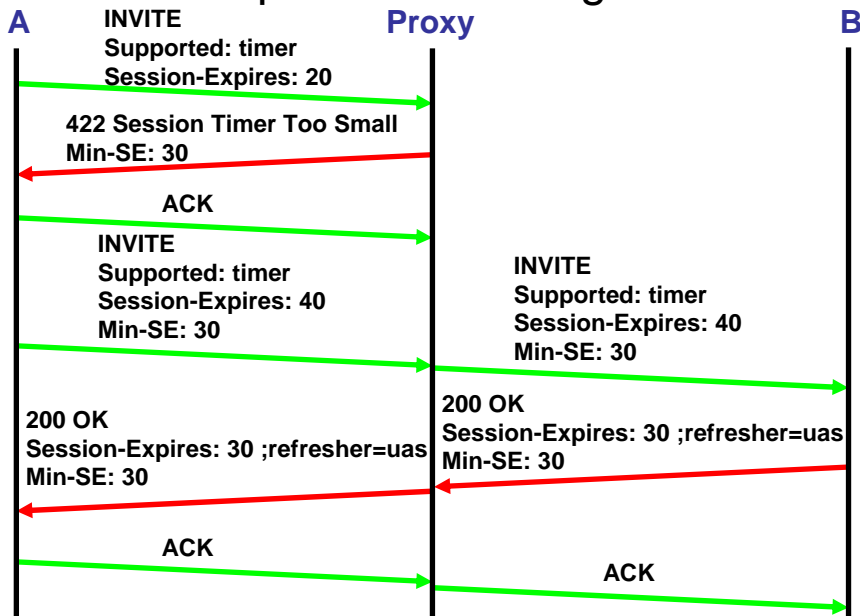
- ▶ No keep-alive mechanism for SIP calls
 - Proxies may preserve outdated state information
 - Close dynamically created holes in firewalls
 - Need timeout-mechanism to perform cleanup
- ▶ SIP Extension for Session timers
 - Option tag *timer*
 - Negotiation mechanism for expiry time and refresh responsibility
 - *Session-Expires*: duration and role
 - 422 response with *Min-SE*: lower bound for expiry interval
 - *Responsible UA* sends re-INVITE before timer expires
- ▶ Session is terminated if no re-INVITE/UPDATE arrives in time
 - *Responsible UA* sends BYE
 - Proxies silently drop state information

Example: Keep-alive



- Caller indicates support of session timers in INVITE
- Propose initial timer value
- Called UA supports timers, accepts proposed value, asks the caller to send refresh
- Setup complete, media streams are established
- After $n/2 \cdot \text{timer}$ calling party sends re-INVITE
- Call continues after 200 OK and ACK

Example: Interval Negotiation

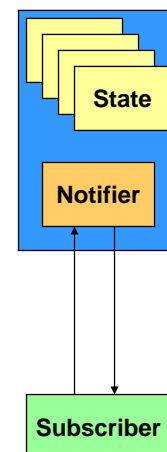


Event Notifications

- ▶ Need for flexible event notification
 - Enable presence information
 - Better support for mobile SIP applications
 - Feedback about progress of other calls, conference state, etc.
 - NOTIFY method, Event:-header
- ▶ Event subscription
 - SUBSCRIBE, Event:
 - Events may be call-related or third-party generated
 - Security issues: Sensitive Events
 - Privacy
 - Authentication
- ▶ Used for personal presence applications
 - Used with PUBLISH method to update presence state
 - Augmented by MESSAGE method for Instant Messaging

Event Concept

- ▶ Piece of state information S
 - Identified by some name (“package”)
- ▶ SIP entities interested in S
 - Query for the current state
 - “polling”
 - Be notified about changes to S
 - SUBSCRIBE
 - Subscriptions may be created implicitly
 - By means of other (SIP or non-SIP) protocol activity
- ▶ Information about S carried in message body
 - NOTIFY
 - Formats to be defined specific to S
- ▶ Protection of S
 - Keep control of who gains access, who has access; for how long





SUBSCRIBE

- ▶ **SUBSCRIBE** used to registered interest in a piece of state
 - **Event**: identifies the state information to be retrieved
 - 489 Bad Event
 - Syntax reflects package concept:
`package ('.' template)*` → e.g. "Event: presence.wininfo"
 - **Allow-Events**: used to indicate which event packages are supported
 - **Expires**: how long to subscribe
 - Subscriptions are soft-state; need to be refreshed periodically
 - May be shortened or lenthened by server
 - **Expires: 0**
 - Poll the state information once; do not establish a subscription
 - Terminate a subscription
 - Body may indicate desired notification policy (filters)
- ▶ Each **SUBSCRIBE** triggers at least one **NOTIFY**
 - May contain no (useful) information if access not yet authorized



SUBSCRIBE Response

- ▶ **SUBSCRIBE Responses**
 - 200 OK: Everything's fine: event understood, client authorized, etc.
 - 202 Accepted : Request received, working on a final result
 - 401, 603, ... if applicable
 - Acceptance or rejection to be reported in NOTIFY
- ▶ **Accepting / rejecting subscription requests**
 - Automated e.g. through configured black / white list
 - Ask user if acceptance cannot be determined automatically
- ▶ **Watcher Info package used to monitor one's own presence**
 - Generates notifications for subscription requests
 - User then needs to authorize subscription
 - Through arbitrary means (e.g. web page)

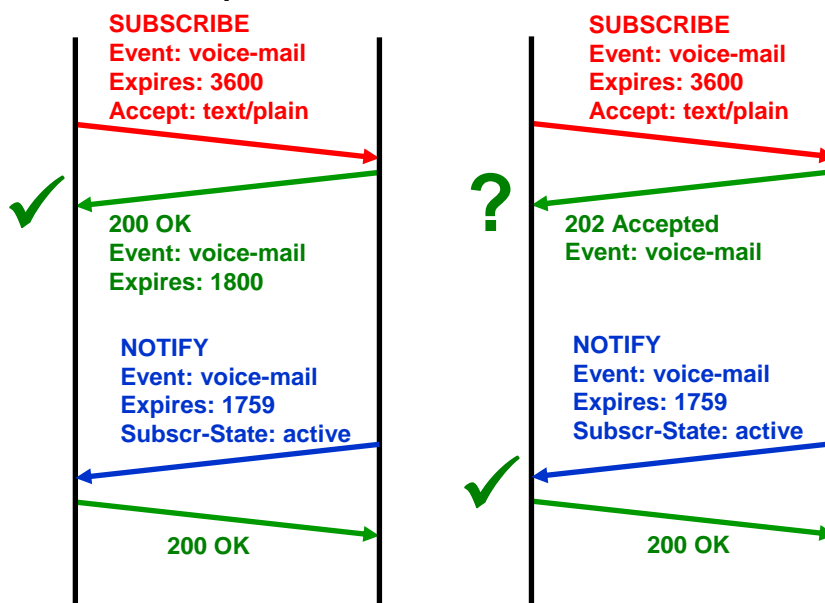


NOTIFY

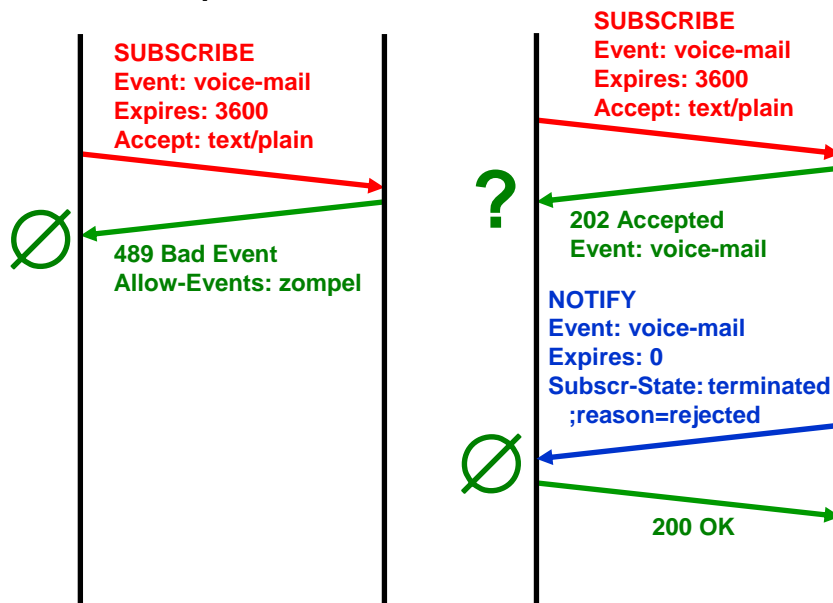
- ▶ NOTIFY carries state information in message body
 - Format depending on **Accept:** header from SUBSCRIBE
 - Full state vs. state deltas
 - **Subscription-State:** active/pending/terminated
- ▶ NOTIFY indicates acceptance or rejection of SUBSCRIBE
 - If no immediate response could be supplied
- ▶ NOTIFY may be used to terminate subscriptions
 - Initiated by the notifier
 - Includes reason code parameter
- ▶ NOTIFY may be sent without SUBSCRIBE
 - Implicit subscription
- ▶ Response to NOTIFY
 - 200 OK vs. 481 *Subscription does not exist*
- ▶ Notification rate: risk of network congestion
 - Event throttling to be specified in event packages



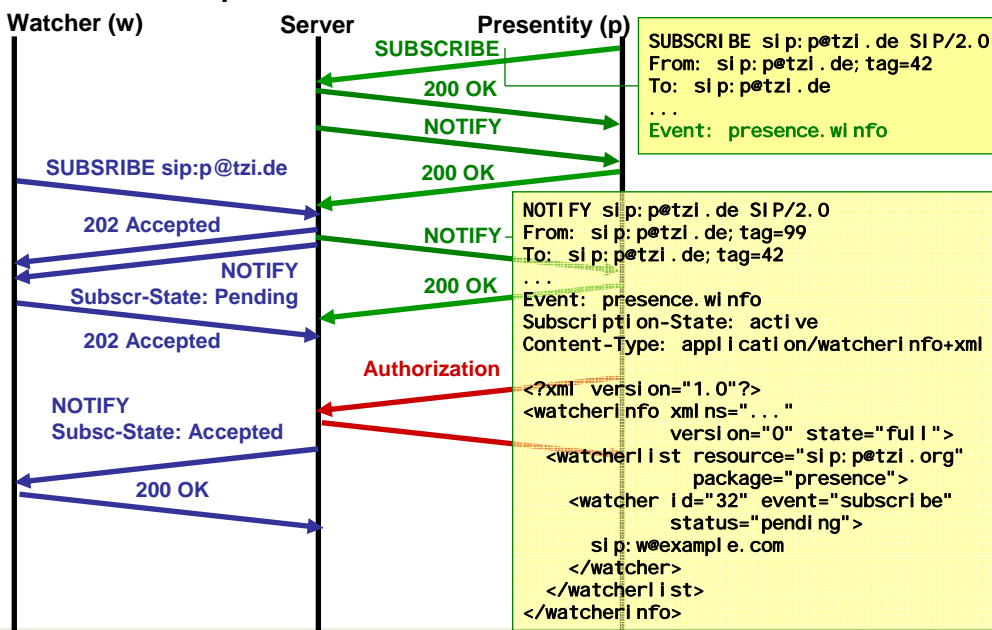
Example: Successful SUBSCRIBE



Example: Unsuccessful SUBSCRIBE



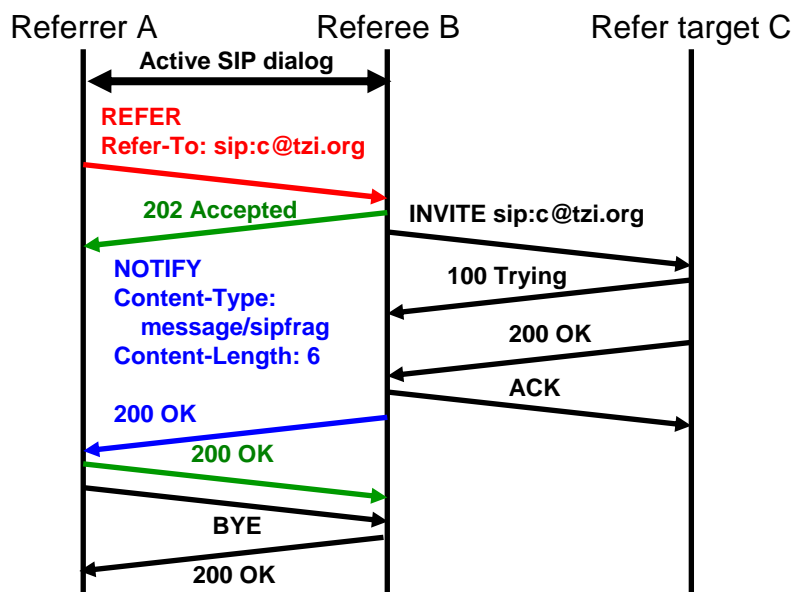
Sample Use of Watcher Information



REFER Method

- ▶ Original motivation: Call Transfer
- ▶ General idea: make a SIP entity contact a third party
- ▶ Originator sends **REFER** method to peer
 - indicates *refer target* in **Refer-To:** header
- ▶ Referred party accepts or declines immediately
 - **202 Accepted** or uses some SIP error code
 - Accepting establishes an *implicit subscription* on the new dialog
- ▶ Contacts *refer target* in an entirely independent dialog
 - May indicate who caused the call in **Referred-By:** header
 - New **INVITE** – **200 OK** – **ACK** sequence, unrelated
- ▶ Reports outcome of new call to originator
 - (exactly) one **NOTIFY**
 - Message body contains SIP message fragments from new call, e.g.
 - **200 OK** **486 Busy Here** **503 Service unavailable**

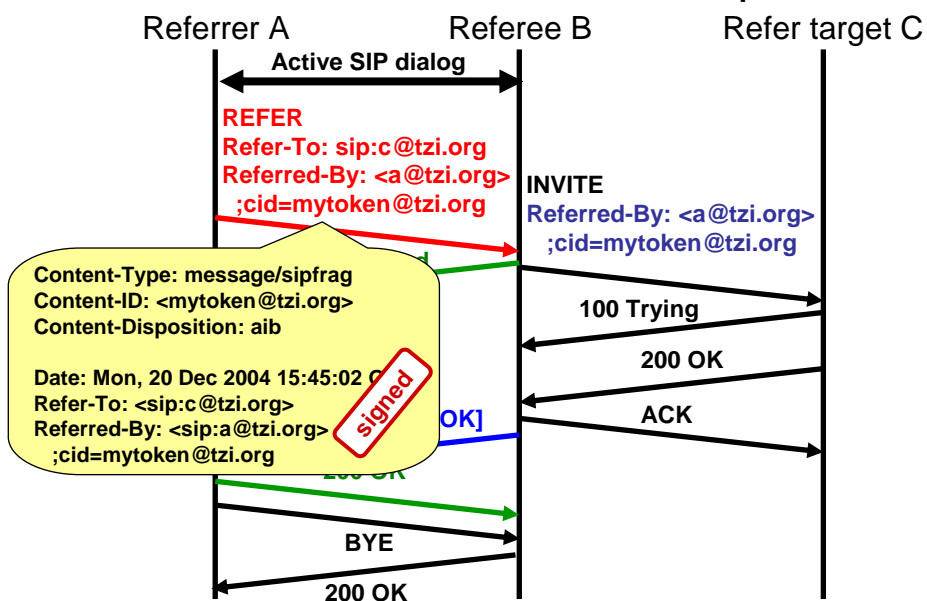
Simple REFER Example



Securing REFER

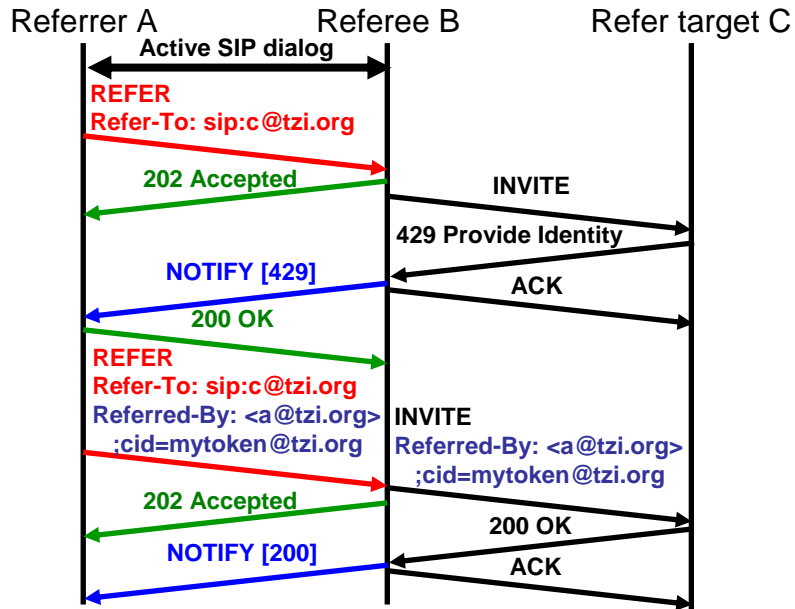
- ▶ Refer Target C receives a request from Referee B
 - May be indistinguishable from any other request from B
 - May be acceptable to C only if A is known as initiator
 - Need to provide authentication information in the request
- ▶ Referred-By: header provides first indication
 - But may be faked by B
- ▶ Extension to securely pass information from A to C
 - Include signed S/MIME message body in REFER request
 - Content-ID: <sip-message-id>
 - To be copied by Referee into target request
 - Optional cid= parameter of Referred-By: points to signed body
 - Referred-By: <sip:referrer@example.com>;cid="<sip-message-id>"
- ▶ Refer target may refuse message without proper Referred-By
 - 429 Provide Referrer Identity

Protected REFER Example





Enforcing Protected REFER Example



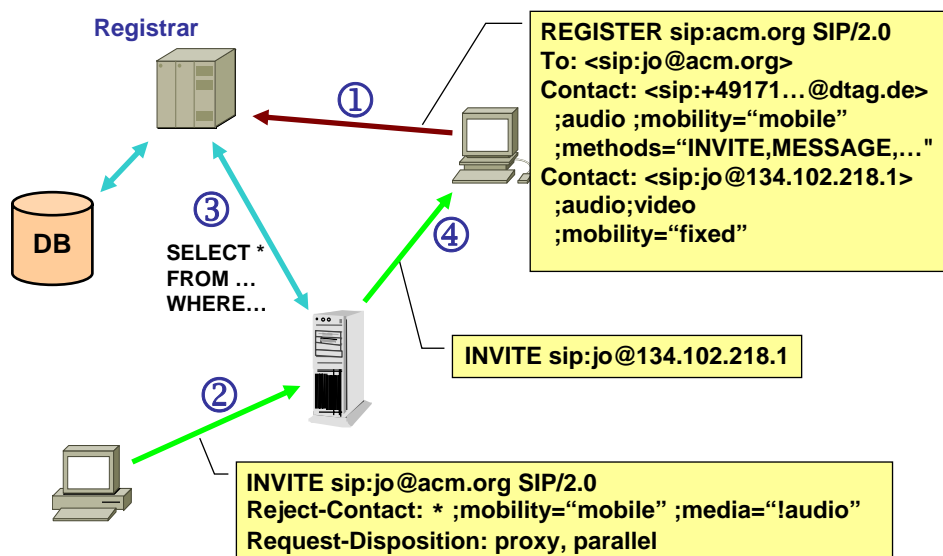
REFER Semantics?

- ▶ Refer-To: supports arbitrary parameters
 - Including `method=` and `response=`
 - May populate various headers of a message
 - Enables remote control of a SIP UA
- ▶ Default: INVITE method for SIP URIs
 - Inside dialog: call transfer
 - Outside dialog: click-to-dial
- ▶ But generalized command: **Apply <method> to <URI>**
 - “Force” a remote UA to do something
 - In principle broad applicability but only little semantics “defined”
 - Fetch the content at `http:// ...`
 - Subscribe to a certain resource
 - Send a certain reply to a request
 - Handle with care...

Caller Preferences / Callee Capabilities

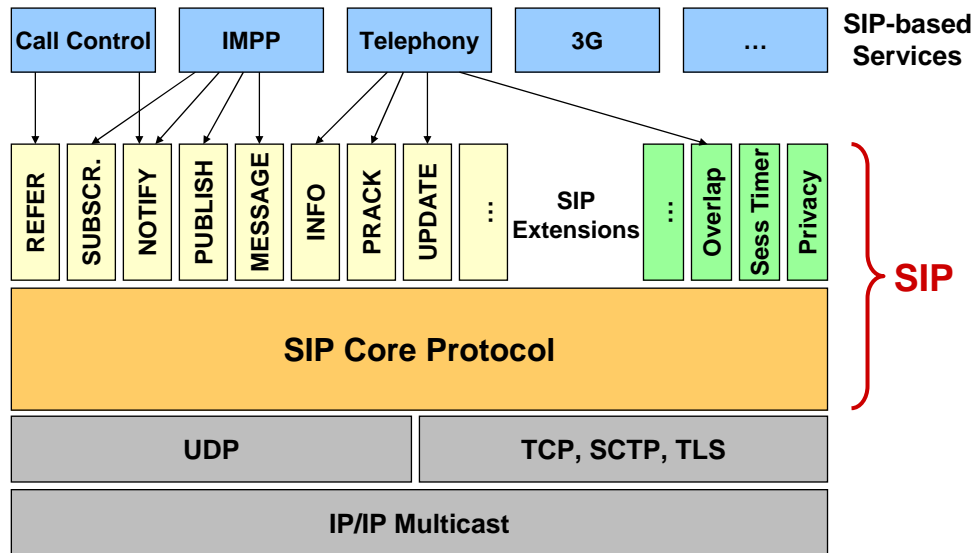
- ▶ Caller may provide preferences for request routing
 - URI-based Proxy or redirect
 - Refuse particular URIs
 - Forking
 - Recursive search
 - Parallel or sequential search
- ▶ SIP Extensions (RFC 3840, 3841)
 - **Request-Disposition:** Request routing in SIP servers
 - **Accept-Contact:** Change address ordering
 - Parameters **require** and **explicit** to control matching process
 - **Reject-Contact:** Reject particular URIs
 - Implied predicates from SIP method, events
 - Additional parameters for Contact: headers in REGISTER
 - Modelled following RFC 2533, but with simplified syntax

Preference/Capability Matching





SIP Service Creation Model



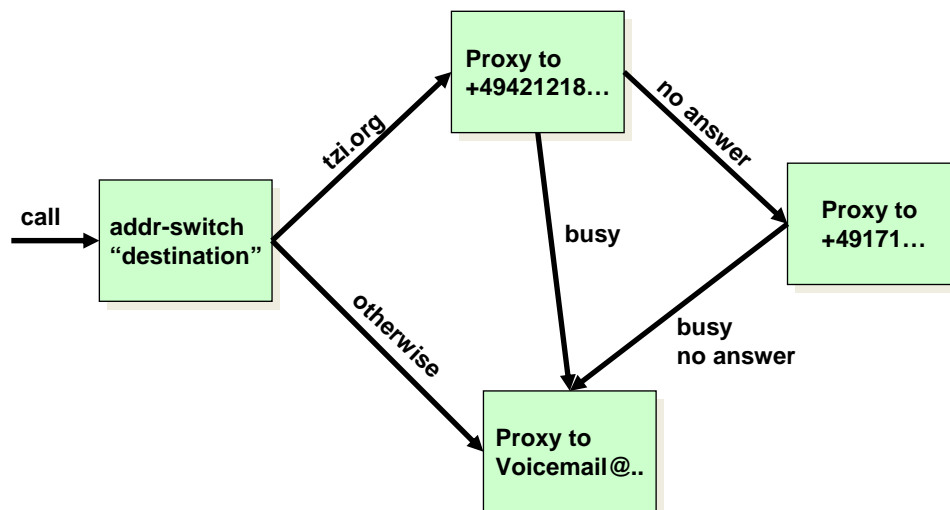
SIP Service Creation

- ▶ Extensions to SIP
 - Methods, headers, protocol handling (as just discussed for some examples)
- ▶ Usages of SIP
 - Combining existing methods, headers, etc. to create a service
 - E.g., SIP for telephony, presence, instant messaging
- ▶ Tuning behavior of SIP entities
 - Based upon information available from SIP
 - Causing effects through SIP mechanisms
 - Accessible by means of APIs: in UAs, proxies, and B2BUAs
 - E.g., user location
- ▶ Interaction with application servers
 - Stimulus signaling without semantic extensions to SIP
 - E.g., using RTP or other protocols to carry signaling

Call Processing Language (CPL)

- ▶ Server provides information about incoming message
 - Destination
 - Originator
 - Caller preferences
 - Contents of several important header fields
 - Media description
 - Security parameters
 - ...
- ▶ CPL scripts can specify several actions to take
 - Reject, redirect or proxy incoming message
 - Set timeout values for actions
 - Context-dependent choice of different actions
 - Perform location lookups
 - ...

Example CPL Script





```
<?xml version="1.0" ?>
<cpl>
  <subaction id="vm">
    <location url="sip:voicemail@dmn.tzi.org">
      <proxy />
    </location>
  </subaction>
  <incoming>
    <address-switch field="destination" subfield="host">
      <address_subdomain-of="tzi.org">
        <location url="tel:+49421218...">
          <proxy timeout="10">
            <busy> <sub ref="vm" /> </busy>
            <noanswer>
              <location url="tel:+49171...">
                [...]
              </noanswer>
            </proxy>
          </location>
        </address>
      </address-switch>
    </incoming>
  </cpl>
```

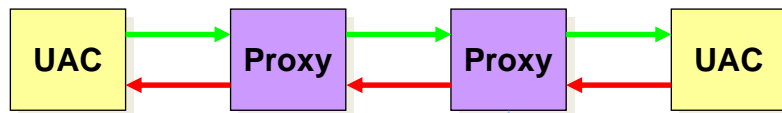


Service Creation

- ▶ Flexible processing of SIP messages in *proxy*
 - Rapid development of supplementary services
 - Pass incoming requests to external script for processing
- SIP Common Gateway Interface (CGI)
 - Adaptation of HTTP CGI
 - Support SIP's idiosyncrasies:
 - Transport protocols UDP, TCP
 - Central role of proxy servers
 - Persistent transaction state
 - Registrations
 - Request forking, recursive search, timeouts
 - ...

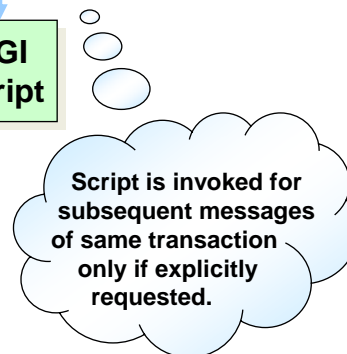
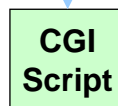


SIP CGI



Adaptation of CGI for HTTP

- Fork script on incoming messages
 - Pass message on STDIN
 - Provide additional information in environment vars
- Process commands from script's STDOUT
 - Message forwarding
 - Create new messages
 - Add/delete message headers
- Use cookies to preserve state information



SIP Service APIs

- ▶ SIP Servlets
 - Standardized model and API for Java-based service creation
 - Built similar to HTTP servlets
- ▶ Vendor-specific APIs
 - for SIP servers
 - SDKs for SIP UAs
- ▶ JAIN (Java APIs for Intelligent Networks)
 - Java-based API for service creation across SIP and circuit-switched networks
 - JAIN SIP
 - JAIN SIP Lite
 - SIP Servlets (see above)
- ▶ Parlay
 - Joint effort of ETSI, 3GPP, and Parlay group

Interaction with Application Services

- ▶ Traditional PSTN / PBX services often controlled via DTMF
 - Stimulus signaling with semantics defined by the application
 - Assumes that signaling and media paths are identical
- ▶ Does not apply to SIP
 - Media and signaling may go separate paths
 - Application servers may be on neither path
- ▶ SIP defines Application Interaction Framework
 - Taxonomy for interactions between users and application servers
 - Focus on stimulus-based signaling
- ▶ KPML Approach
 - SUBSCRIBE to register interest of an application server in user input
 - XML document to describe input expected from user
 - NOTIFY to inform server about key events

Signaling-based Application Servers

