SIP in Mobile Applications

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Overview

- mobility more than just wireless terminals
- SIP for mobility
- SIP bake-off

Mobility in an IP Environment

Terminal mobility: terminal moves between subnets

Personal mobility: different terminals, same address

Service mobility: keep same services while mobile

Terminal Mobility

- domain of IEEE 802.11, 3GPP, mobile IP, ...
- main problems:
 - handover performance
 - handover failure due to lack of resources in new network
 - authentication of redirection

Personal Mobility

- switch between PDA, cell phone, PC, Ethernet phone, Internet appliance, ...
- several "generic" addresses, one person/function, many terminals
- e.g., tel:2129397042, hgs@cs.columbia.edu, schulzrinne@yahoo.com or support@acme.com
- SIP is designed for that proxying and redirection does translation
- but: need mapping mechanisms to recognize registrations as belonging to the same person
- some possible solutions:
 - dip into LDAP personnel database or /etc/passwd to match phone number and variations of name (*J.Doe*, *John.Doe*, *Doe*)
 - need dialing plan to recognize 7042@cs.columbia.edu and tel:2129397042 as same

Service mobility

Examples:

- speed dial & address book
- media preferences
- special feature buttons (voice mail, do-not-disturb)
- incoming call handling instructions
- buddy lists
- → independent of terminal (including pay phone!), across providers

Service mobility

- REGISTER can retrieve configuration information (e.g., speed dial settings, distinctive ringing or voice mail settings)
- but needs to be device-independent
- most such services (e.g., voicemail forwarding, call filtering) should remain on server(s)

Separate issue: how does the payphone (or colleague's phone) recognize you?

- PDA (IR)
- i-button
- fingerprint
- speech recognition, ...

One device, but changing set of owners!

Service mobility – call handling

- need uniform basic service description model → Call Processing Language (CPL)
- CPL = XML-based flow graph for inbound & outbound calls
- CPL for local call handling
- update CPL from terminal: add telemarketer to block list
- harder: synchronize CPL changes across multiple providers
- one possibility: REGISTER updates information, but device needs to know that it has multiple identities
- merging of call logs

Terminal Mobility – Details

move to new network IP address changes (DHCP)

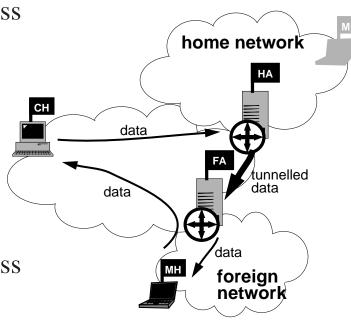
mobile IP hides address changes

• but: little deployment

• encapsulation overhead

dog-legged routing

may not work with IP address filtering



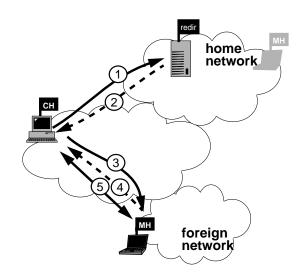
- mobile host
- ch correspondent host
- router with home agent functionality
- router with foreign agent functionality

SIP mobility overview

- pre-call mobility SIP proxy, redirect
- mid-call mobility SIP re-INVITE, RTP
- recovery from disconnection

SIP mobility: pre-call

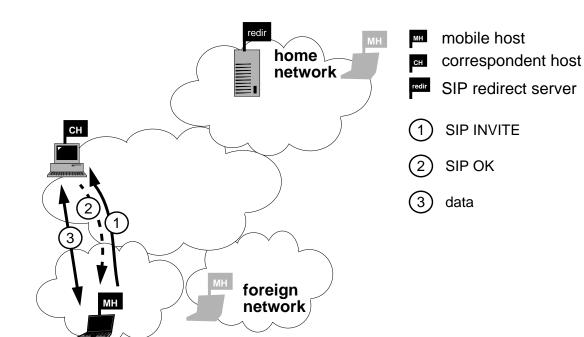
- MH acquires IP address via DHCP
- optional: MH finds SIP server via multicast REGISTER
- MH updates home SIP server
- optimization: hierarchical LR (later)



- mobile host
- ch correspondent host
- redir SIP redirect server
- (1) SIP INVITE
- 2 SIP 302 moved temporarily
- (3) SIP INVITE
- (4) SIP OK
- (5) data

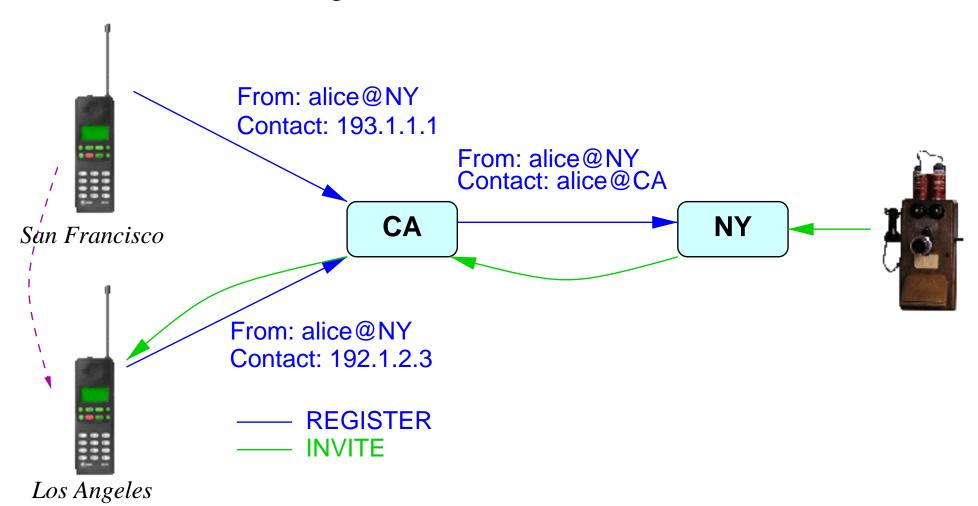
SIP mobility: mid-call

- MH→CH: new INVITE, with Contact and updated SDP
- re-registers with home registrar



SIP mobility: multi-stage registration

Don't want to bother home registrar with each move



SIP and mobility: issues

- doesn't work for TCP applications solutions:
 - punt: "don't walk while telnet'ing"
 - application-layer awareness: restart web, email, ftp transfer need for deep fade anyway...
 - NAT-style boxes controlled by SIP (see Telcordia ITSUMO project)
- but: works nicely for "vertical handoff" between different technologies e.g., transfer call from mobile handset to office videophone when arriving at work

Conclusion

- mobility is more than just wireless handsets
- terminal, personal and service mobility
- SIP enables all three, but likely to be hybrid solutions

For more information...

SIP: http://www.cs.columbia.edu/sip

RTP: http://www.cs.columbia.edu/~hgs/rtp

Papers: http://www.cs.columbia.edu/IRT