SIP for Mobile Applications

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

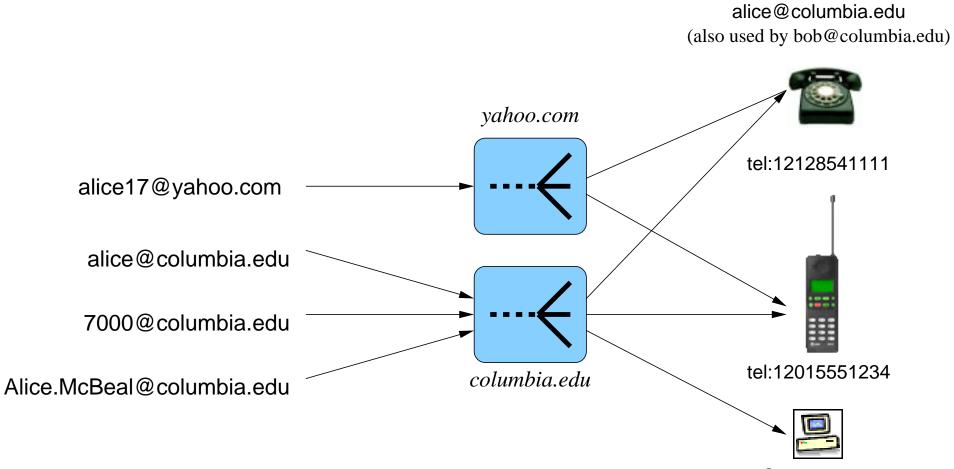
Mobility in an IP environment

Terminal mobility: terminal moves between subnetsPersonal mobility: different terminals, same addressService 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



alice@host.columbia.edu

- 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
- \rightarrow independent of terminal (including pay phone!), across providers

- 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, ...

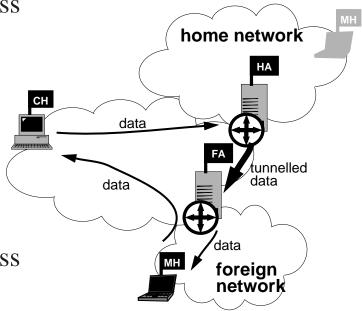
One device, but changing set of owners!

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



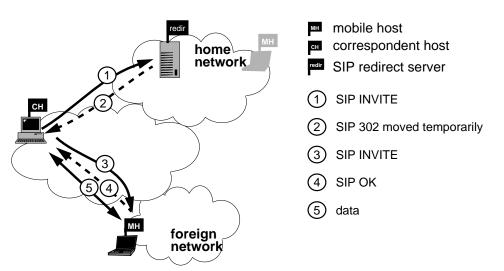
- mobile hostcorrespondent host
- HA router with home agent functionality
- router with foreign agent functionality

SIP terminal mobility overview

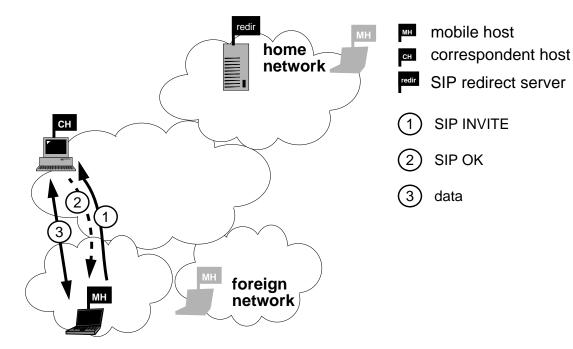
- pre-call mobility IP proxy, redirect
- mid-call mobility IP re-INVITE, RTP
- recovery from disconnection

SIP terminal 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)



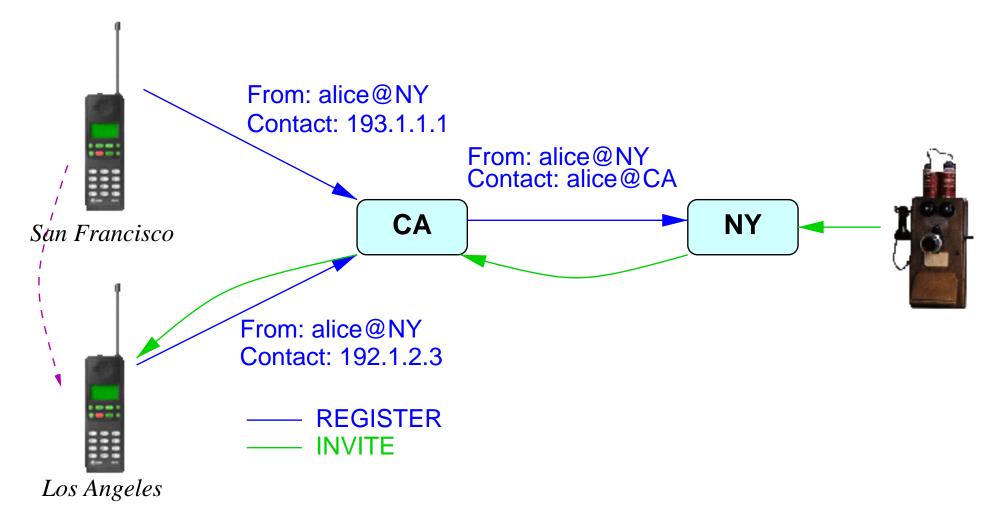
SIP terminal mobility: mid-call



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

SIP terminal 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

- 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