## Internet Telephony for Universities

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## Overview

- Internet telephony: motivation and problems
- Campus VoIP architectures
- Session Initiation Protocol (SIP)
- Internet telephony "appliances"
- Programming your telephone (service)
- Mobile services


## The phone works - why bother with VoIP?

user perspective

- variable compression: tin can to broadcast quality
- security through encryption
- caller, talker identification
- better user interface
- internat. calls: TAT transatlantic cable $=\$ 0.03 / \mathrm{hr}$
- no local access fees (3.4c)
- easy: video, whiteboard, ...
- silence suppression III traffic $\downarrow$
- shared facilities management, redundancy
- advanced services (simpler than AIN and CTI)
- operational advantages
- cheaper switching
- fax as data


## The new phone companies

- separation bit carriage $\leftrightarrow$ services
- anybody with Internet connection can provide services (ACD, 800, 900, directory, ...)
- distinction "in" vs. "out" of network not useful
- incremental start-up investment not large
- new players:
- cable companies II* no new infrastructure, but mostly one-way
- electric utilities
- Qwest, IXC (resell to ISPs), ...


## Internet telephony services

- voice mail $\longrightarrow$ email
- calendar integration
- user-programmable call processing logic
- call first available sales person (ACD)
- call whole department
- web IVR
- return web page with favorite "on hold" music


## Internet Telephony Services

- camp-on without holding a line
- short message service ("instant messaging")
- schedule call into the future
- call with expiration date
- add/remove parties to/from call ${ }^{\text {n+ }}$ mesh
- "buddy lists"


## Switching Costs

| Device | port speed | port cost | cost/64 kb/s |
| :--- | ---: | ---: | ---: |
| 8-port Ethernet hub | $10 / 100 \mathrm{Mb} / \mathrm{s}$ | 8 | 0.008 |
| 24-port Ethernet switch | $10 \mathrm{Mb} / \mathrm{s}$ | 55 | 0.35 |
| 8-port Ethernet switch | $100 \mathrm{Mb} / \mathrm{s}$ fiber | 474 | 0.30 |
| 8-port Ethernet switch | $1 \mathrm{~Gb} / \mathrm{s}$ | 1187 | 0.08 |
| 24×100BaseT + GigE | $10 / 100 \mathrm{Mb} / \mathrm{s}$ | 141 | 0.09 |
| 100 T1 circuit switch | $1.5 \mathrm{Mb} / \mathrm{s}$ | 25,000 | 1041 |
| 5ESS local (no AIN), 5000 lines | $64 \mathrm{~kb} / \mathrm{s}$ | 300 | 300 |
| 5ESS local (AIN), 20,000 lines | $64 \mathrm{~kb} / \mathrm{s}$ | 175 | 175 |
| Small PBX (few hundred lines) | $64 \mathrm{~kb} / \mathrm{s}$ | 1,000 | 1,000 |
| Large PBX (>5000 lines) | $64 \mathrm{~kb} / \mathrm{s}$ | 500 | 500 |

## Telephone Costs



## Transport Costs

| network | $\$ / \mathrm{min}$ | $\$ / \mathrm{MB}$ |
| :--- | :--- | :--- |
| wholesale telephone | $0.01-0.02$ |  |
| U.S. domestic interstate consumer rates | $0.05-0.15$ |  |
| U.S. domestic intrastate consumer rates | $0.05-0.25$ |  |
| modem |  | $0.25-0.50$ |
| private line | $0.50-1.00$ |  |
| frame relay | 0.30 |  |
| MCI frame SVC | 0.05 |  |
| Internet | $0.04-0.15$ |  |
| Internet modem | 0.33 |  |
| Internet backbone | 0.01 |  |

$1^{\prime}$ voice $=480 \mathrm{kB} \mathrm{w} /$ silence suppr., 1 MB without

## Phone Usage

"Free" phone calls does not mean unbounded increase:

| year | lines <br> (millions) | local calls <br> $\mathrm{min} /$ day/line | local calls <br> $\mathrm{min} /$ day/person |
| ---: | ---: | ---: | ---: |
| 1980 | 102.2 | 39 | 17.5 |
| 1988 | 127.1 | 39 | 20.2 |
| 1996 | 166.3 | 40 | 25.1 |

## Why Aren't We Using It Now?

Internet capacity $\ll$ phone traffic:

| world phone traffic | 600 | $\mathrm{~Gb} / \mathrm{s}$ | U.S. total | 368 | $\mathrm{~Gb} / \mathrm{s}$ |
| :--- | ---: | :--- | :--- | ---: | :--- |
| international traffic | 13 | $\mathrm{~Gb} / \mathrm{s}$ | U.S. interstate | 55 | $\mathrm{~Gb} / \mathrm{s}$ |
|  |  |  | AT\&T long distance | 61 | $\mathrm{~Gb} / \mathrm{s}$ |

- unpredictable sound quality, reliability
- doesn't work well for dial-up users
- no cheap Internet devices
- 640 M phone lines, 122 M in U.S.
- no billing infrastructure


## Projections

- MCI: " $80 \%$ data, $20 \%$ voice"
- "AT\&T could lose $\$ 350$ million in international calls by 2001"
- "By 2002, the Internet could account for $11 \%$ of U.S. and international long-distance voice traffic"
- "Up to $10 \%$ of the world's fax market, which generates $\$ 45$ billion in telecom revenue a year, will move to Internet in 2 or 3 years"
- May 1999: BT builds IP phone network in Spain
- but: cable modems only 250,000 to 275,000 users in US, $10 \%$ of Internet users by 2000


## Data vs. Voice Traffic



## Why on Campus?

- PBX nearing end of useful life, capacity
- dorm rooms, offices already wired with Cat-3/5
- backbone high-speed data capacity (20,000 users at 0.1 Erlang $\rightarrow 128 \mathrm{Mb} / \mathrm{s}$, but not all calls are across campus)
- no latency issues
- video, data sharing
- re-use data connections as tie-lines to satellite campuses, dorms, faculty housing, ...


## Internet Telephony Architecture



## Campus Data Architecture



## Architecture Options

- separate wiring vs. same network
- stimulus control vs. intelligent end systems
- IP Centrex vs. external PSTN interface


## A Campus VoIP Architecture



## Separate Wiring

- re-use CAT3 wiring $\longrightarrow$ just requires centralized changes
- but: distance limitation of $100-150 \mathrm{~m}$
- power requirements:

Etherphones
Wireless access point
Ethercams
Ethernet hub

- powering for end systems and hubs:
- local battery
- Ethernet powering


## Ethernet Power

Ethernet cable (802.3 working group):

- phantom powered on $3 / 6,1 / 2$
- idle wires (4/5 and/or 7/8)
- automatic recognition of powered devices

Do all systems need to be powered?

## Architecture for 20,000 Lines



## Stimulus Control vs. Intelligent End Systems

|  | stimulus | end system |
| :--- | :--- | :--- |
| protocol | MGCP | SIP, H.323 |
| $>1$ service provider | no | yes |
| new services | upgrade MGC | proxy, end system software |
| user interface | like phone | more state information |
| scaling | single server | distributed |
| simple devices | yes | SIP: yes, H.323:? |

## Quality of Service

- codecs can be same or better than POTS
- primarily, delay:
audio encoding/decoding: look-ahead, block (20-50 ms)
application: non-adaptive playout buffers
end system: operating system, sound card (buffer) propagation: $5 \mu \mathrm{~s} / \mathrm{km}$
queueing: depends on congestion
transmission: line speed; insignificant for $\geq \mathrm{T} 1$


## Delay

- ITU.T delay target $<150 \mathrm{~ms}$
- average vs. peaks!
- avg. US round-trip (UUnet, Oct. 1999): 45.49 ms
- Miami - Seattle (CWI, Nov. 1999): 92.4 ms

