

Myriad of different LAN technologies co-existing in a WAN. For example:

- Fast Ethernet (100 Mbps)
- Gigabit Ethernet (1000 Mbps); 10 and 100 GigE
  - Purdue CS backbone: 10 Gbps
  - AT&T (tier-1 provider): 10+ Gbps
- WLAN (11, 54, 300 Mbps)
- 4G cellular (100 Mbps mobile, 1 Gbps stationary)
  - today: pre-4G
  - ITU-R
- WiMAX (tens of Mbps up to 1 Gbps)
  - wider area: several miles
- modem/DSL (cable and dial-up)
  - 12 Mbps (down)/2 Mbps (up), 50/10, higher

Note: WAN is a collection of LANs

→ ultimately: everything happens at LANs

Each LAN, in general, speaks a different language

→ message format (syntactic)

→ behavioral (semantic)

Internetworking handles this problem by translating everything to IP (Internet Protocol)

→ technical definition of **I**nternet

→ collection of interconnected LANs speaking IP

But:

- IP injects overhead
- packet forwarding example: switches may choose not to speak IP

Hence:

- IP (layer 3) is not necessary
- large systems of layer 2 (LAN) switches
- size limit: heterogeneity and management headache
- today: L2 + L3
  - common attitude: avoid IP if possible
  - most devices speak IP in case needed
  - IP provides management benefit: naming

IP provides naming flexibility:

- IP: v4 32-bit, v6 128-bit
- in addition to 48-bit LAN addresses are hardwired and unique address per NIC
- IP provides additional configurability

Common practice: assign similar addresses to network devices belonging to same organization

→ ARIN in the U.S.

→ blocks of contiguous addresses: makes routing easier

→ e.g.: Purdue 128.10.\*.★, 128.210.\*.★

→ LWSN B158: sslab01.cs.purdue.edu 128.10.25.101

→ CS web server: www.cs.purdue.edu 128.10.19.20

→ router bottleneck: table look-up speed

Another configurability benefit:

→ private addresses

→ e.g., 10.\*.\*.\*, 192.168.\*.\*

→ laptop WLAN IP address in LWSN: 10.184.43.63

→ no one outside Purdue can reach laptop using 10.184.43.63

→ how is it solved?

Naming: IP or LAN addresses are not enough

Communicating entities are *processes* running on host/router operating systems (Linux, Windows, IOS, etc.)

→ IP only specifies host/server/router

→ more accurately: one of the NICs attached to a device

→ host with multiple NICs: multiple IP addresses

→ multi-homed

Hence:

A name/address must also identify which process a message is destined for on a host

→ OS/network convention: port number abstraction

→ 16-bit

→ address: pair (host IP, port number)

→ well-known server port numbers (e.g., 80 for HTTP)

→ is client app's port number important?



## Network Performance

In computer networks, speed is at a premium

- if slow, typically not used in practice
- e.g., cryptographic protocols tend to be not turned on at routers
- emphasis on lightweight network core
- push heavyweight stuff toward the edge (i.e., host/server)
- called end-to-end paradigm
- has guided Internet design and evolution

Three yardsticks of performance:

- bandwidth: bps (bits-per-second)
  - throughput: includes software processing overhead
  - e.g., 802.11b WLAN: nominal bandwidth 11 Mbps, throughput around 6 Mbps
- latency: msec (millisecond)
  - signal propagation speed
  - approximately: speed of light
  - delay: includes software processing overhead and waiting time at routers (queueing)
  - delay at high speed routers: very small ( $\mu$ sec)
  - delay at WLAN AP: up to hundreds of millisecond
- jitter: delay variation
  - not good for real-time content (video, audio, voice)

Bandwidth vs. throughput:

*bandwidth*—maximum data transmission rate achievable at the hardware level; determined by signalling rate of physical link and NIC

*throughput*—maximum data transmission rate achievable at the software level; overhead of network protocols inside/outside OS is accounted for

*reliable throughput*—maximum reliable data transmission rate achievable at the software level; effect of recovery from transmission errors and packet loss accounted for

→ networks tend to be “leaky”

Meaning of “high-speed” networks:

- signal propagation speed is bounded by SOL (speed-of-light)
  - $\sim 300\text{K km/s}$  or  $\sim 186\text{K miles/s}$
  - optical fiber, copper: nearly same
- Ex.: latency: Purdue to West Coast
  - around 2000 miles:  $\sim 10\text{ msec}$  ( $= 2000/186000$ )
  - lower bound
- Ex.: geostationary satellites:  $\sim 22.2\text{K miles}$ 
  - latency:  $\sim 120\text{ msec}$
  - end-to-end (one-way):  $\sim 240\text{ msec}$
  - round-trip (two-way):  $\sim 480\text{ msec}$
  - typically:  $\sim 500\text{ msec}$

- thus: a single bit cannot go faster
  - can only increase “bandwidth”
  - analogous to widening highway, i.e., more lanes
  - simultaneous transmission of multiple bits
  - hence “broadband” is a more accurate term
- interpretation: “high-speed”  $\Leftrightarrow$  “many lanes”
  - what does it buy?
  - completion time of large files faster
  - in this sense, “higher” speed

Some units:

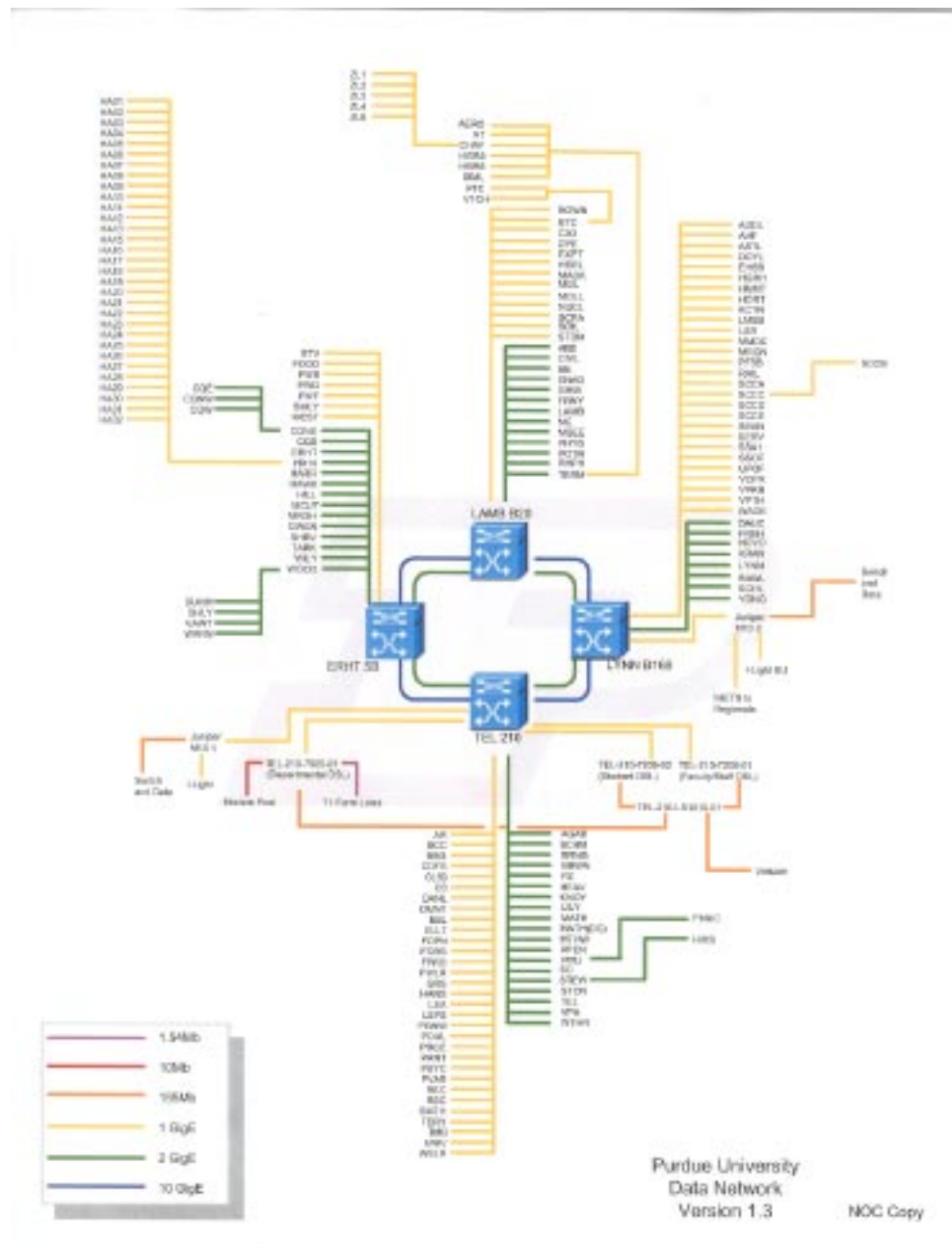
Tbps, Gbps, Mbps, Kbps:

$10^{12}$ ,  $10^9$ ,  $10^6$ ,  $10^3$  bits per second; indicates data transmission rate; influenced by clock rate (MHz/GHz) of signaling hardware

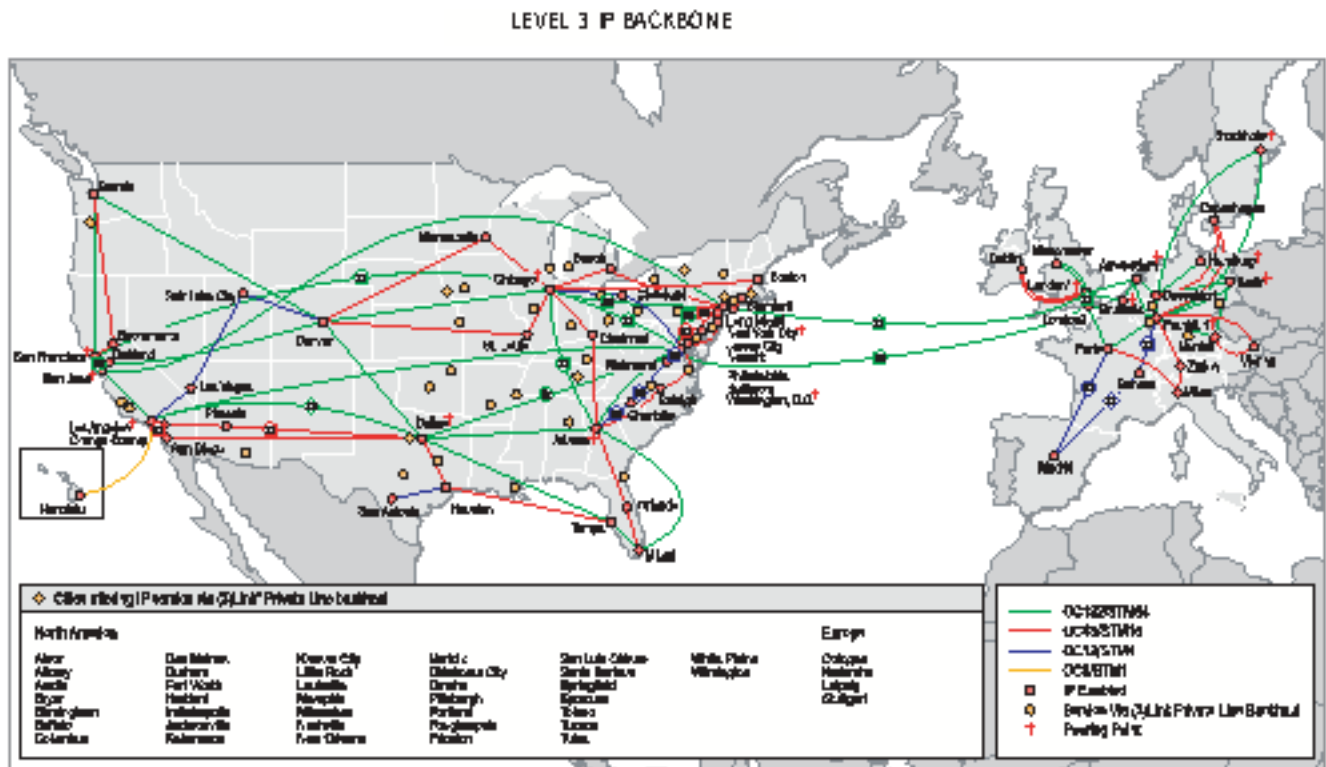
→ communication rate: factors of 1000

→ data size: 1 KB means 1024 bytes

## Purdue's backbone network: ITaP



Level3 backbone network: [www.level3.com](http://www.level3.com)



→ 10 Gbps backbone (green): same speed as Purdue

→ outdated pic: faster backbone speeds now



## What is traveling on the wires?

Mixture of:

bulk data (data, image, video, audio files), voice, streaming video/audio, real-time interactive data (e.g., games), etc.

→ around 90% of Internet traffic is TCP file traffic

→ HTTP web and P2P

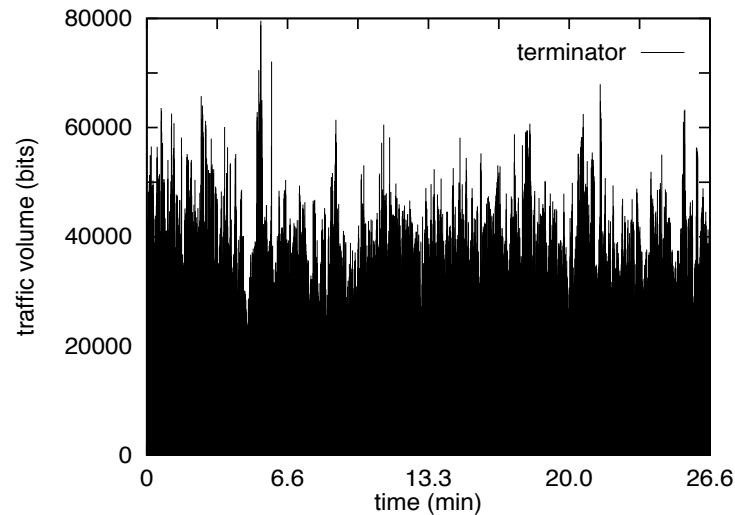
Multimedia (video/audio) streaming: on the rise

→ a minority but share is increasing

→ non-real-time: e.g., youtube, netflix

→ real-time: e.g., VoIP, video conferencing, games

Internet traffic is “bursty”: MPEG compressed real-time video



Reason:

- video compression
  - utilize inter-frame compression
- across scenes, significant scenary changes
  - e.g., action movies
- within scenes, few changes

Burstiness is not good for networks: why?

Main source of traffic burstiness:

→ skewed file size

- 90/10 rule

- 90% of files are small, 10% are very large

→ “many mice, few elephants”

→ the few elephants make up 90% of total traffic

→ same for disk space

Real-world is inherently skewed ...

## How to make sense of all this?

Study of networks has three aspects:

- architecture
  - system design, real-world manifestation
- algorithms
  - how do the components work
- implementation
  - how are the algorithms implemented

Key concern: performance

→ speed

→ increasingly important: security and reliability