Chapter 1: roadmap

1.1 What *is* the Internet?

1.2 Network edge

- ❑ end systems, access networks, links

1.3 Network core

- ❑ network structure, circuit switching, packet switching

1.4 Delay, loss and throughput performance in packet-switched networks

1.5 Protocol layers, service models

1.6 Networks under attack: security

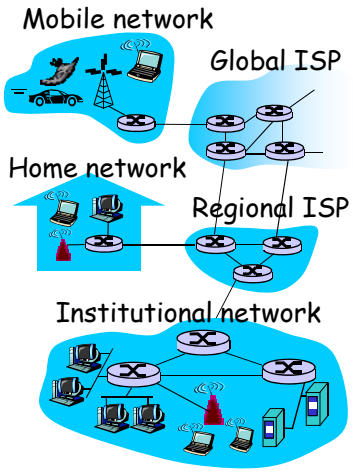
1.7 History

What's the Internet: "nuts and bolts" view

- ❑ billions of connected computing devices: *hosts = end systems*
  - ❖ running *network apps*

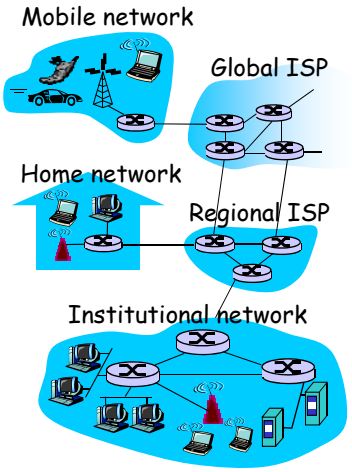
- ❑ *communication links*
  - ❖ fiber, copper, coax, radio, satellite
  - ❖ transmission rate

- ❑ *Routers and switches* forward packets



What's the Internet: architecture & protocols

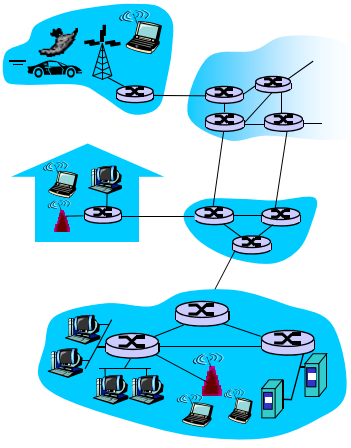
- *Internet*: "network of networks"
  - loosely hierarchical
  - public Internet versus private intranet
- *protocols* control sending, receiving of msgs
  - e.g., TCP, IP, HTTP, BGP, Ethernet, Skype
- Internet standards
  - RFC: Request for comments
  - IETF: Internet Engineering Task Force



What are required for global connectivity?

What's the Internet: a service view

- communication *infrastructure* enables distributed applications:
  - Web, VoIP, email, games, e-commerce, file sharing
- communication services provided to apps:
  - reliable data delivery from source to destination
  - "best effort" (unreliable) data delivery



What's a protocol?

human protocols:

- "what's the time?"
- "I have a question"
- introductions

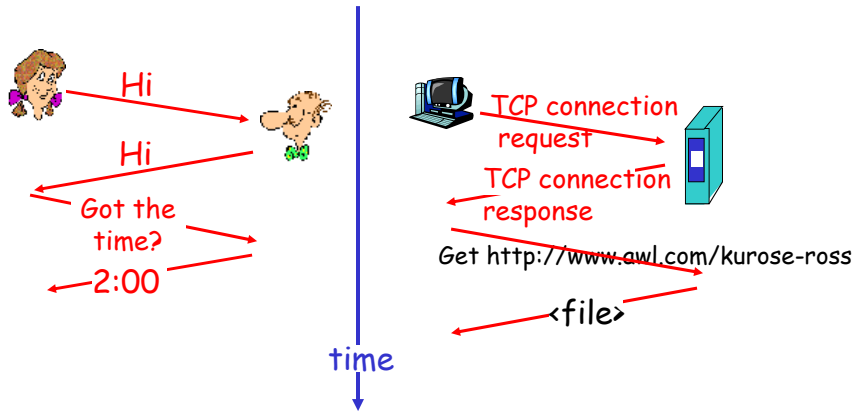
network protocols:

- machines rather than humans
- all communication activity in Internet governed by protocols

protocols define format, order of msgs sent and received among network entities, and actions taken on msg transmission, receipt, or timeout

What's a protocol?

a human protocol and a computer network protocol:

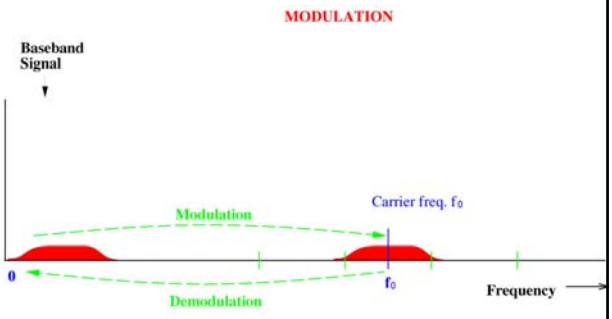


Q: Other human protocols?

From physical media to communication channels—basic concepts (not in textbook)

Modulation and Demodulation

- Common examples: radio, television channels for analog signals
  - Bandwidth in hertz



- Can also be used for digital signals (encoding binary data)

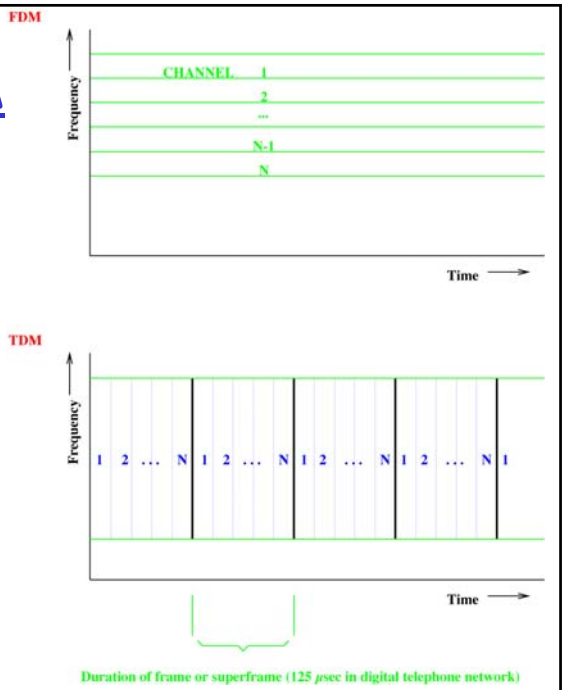
$$A\cos(2\pi f_0 t+\theta)$$

Shannon's Theorem

$$C = B \log_2 (1 + S/N)$$

where
- C max capacity in bits/sec
- B bandwidth in hertz
- S/N signal to noise ratio

FDM vs. TDM



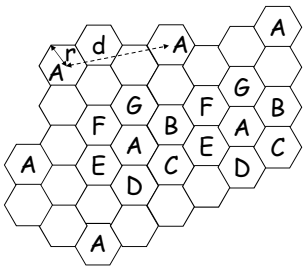
Duration of frame (or superframe) is 125 μsec in digital telephone networks

TDM in Telephone Networks

- Why 125 μsec for frame duration?
- Sampling Theorem: An analog signal can be reconstructed from samples taken at a rate equal to twice the signal bandwidth
- Bandwidth for voice signals is 4 Khz; for hi fidelity music, 22.05 Khz per channel
- Sampling rate for voice = 8000 samples/sec or one voice sample every 125 μsec
- Digital voice channel (uncompressed), 8 bits x 8000/sec = 64 Kbps

Other Multiplexing Techniques

- Space division multiplex
  - Same frequency used in different cables
  - Same frequency used in different (nonadjacent) cells



- Wavelength division multiplex
  - Light pulses sent at different wavelengths in optical fiber
- Code division multiplex (in chapter 7 of text)

e.g., CDMA for cell phones

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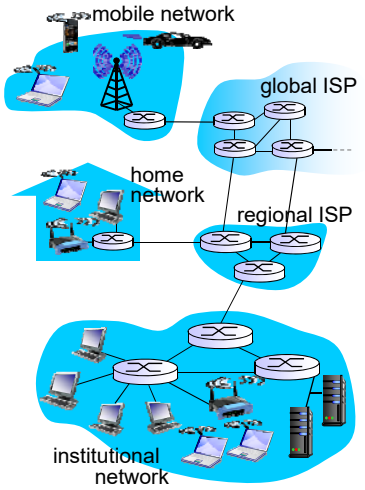
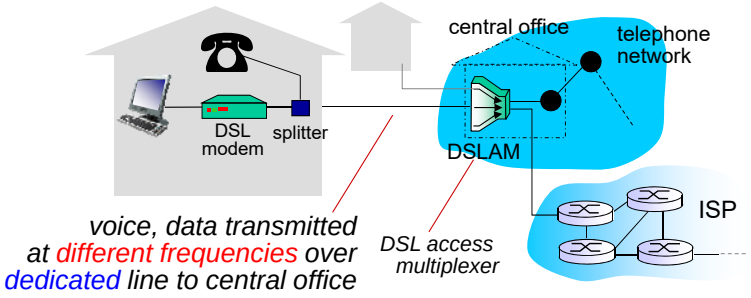
A closer look at network structure:

- *network edge:*
  - hosts: clients and servers
  - servers often in data centers
- *access networks, physical media:* wired, wireless communication links
- *network core:*
  - interconnected routers
  - network of networks

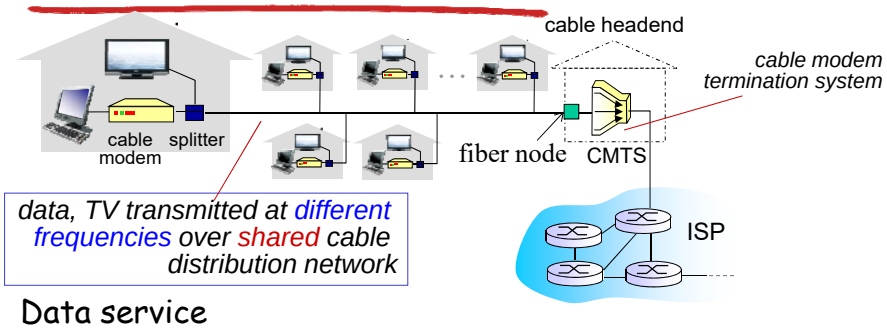


Access net: digital subscriber line (DSL)



- use FDM in telephone line to central office DSLAM
  - data over DSL line goes to Internet
  - voice over DSL line goes to telephone net
- asymmetric bandwidths/transmission rates (data download much faster than upload)

Access net - hybrid fiber coax (HFC)

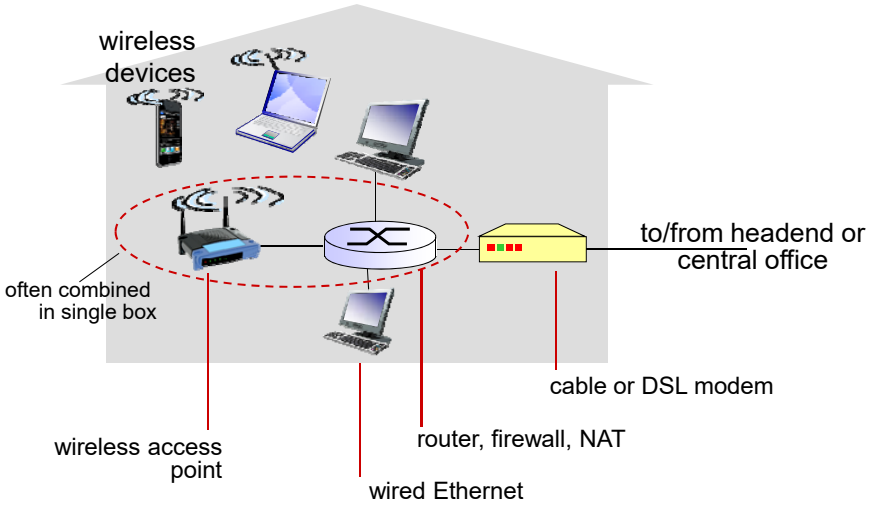


Data service

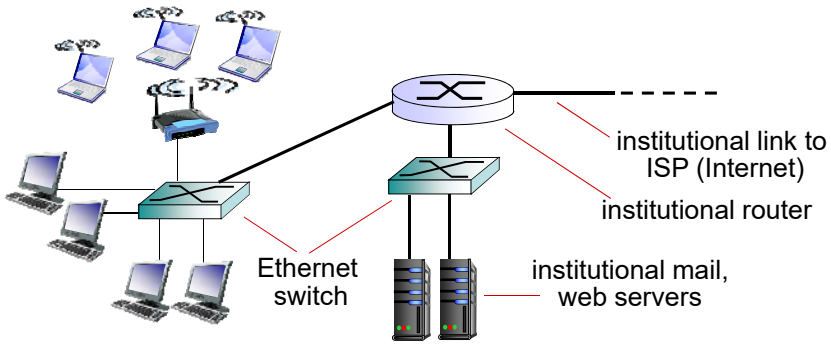
- homes *share* coax cable to cable headend
  (*unlike DSL*, which has dedicated access to central office)
- data channels have asymmetric rates and they are ***shared*** by homes - *multiple access protocol* required for **uplink**

Fiber to the home (Verizon, Google) – all optical switches

Access net: home network



Enterprise access networks (Ethernet)



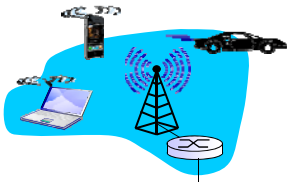
- today, end systems typically connect into Ethernet switch
  - 100Mbps, 1Gbps, 10Gbps transmission rates
- A large enterprise network is connected to multiple ISPs
  - multi-homing

Wireless access networks

- shared *wireless* access network connects end system to router
  - via "access point" or "cell tower"

wireless LANs:
- within building (100 ft)
- 802.11g/n/ac (WiFi)

wide-area wireless access
- provided by telco (cellular) operators, 10's km (when no obstacle)
- 3G, 4G: LTE



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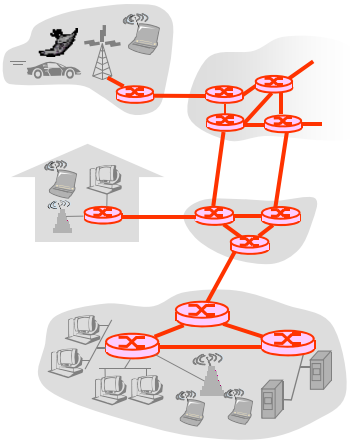
The Network Core

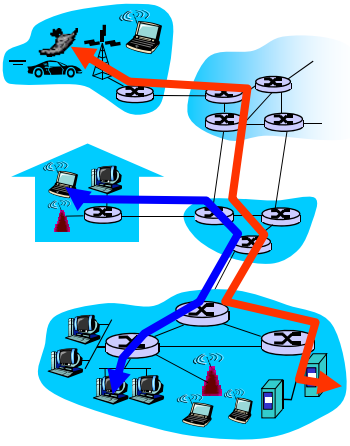
- mesh of interconnected routers
- *the* fundamental question: how is data transferred through net?
  - circuit switching: dedicated circuit per call: telephone net
  - packet-switching: data sent thru net in discrete "chunks"



Network Core: Circuit Switching

End-to-end resources reserved for each "call"

- E.g., link bandwidth
  - FDM, TDM
- end-to-end circuit-like (guaranteed) performance
- call setup required
  - resource piece *idle* if not used by the call *(no sharing)*



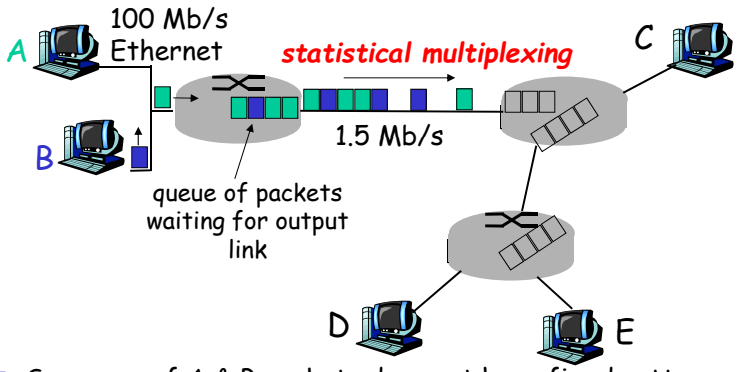
Numerical example

- How long does it take to send a file of 640,000 bits from host A to host B over a *circuit-switched* network?
  - all links are 1.536 Mbps
  - each link uses TDM with 24 slots/sec (i.e., one slot per circuit)
  - 500 msec to establish end-to-end circuit

Let's work it out!

Note: 1.536 Mbps was realistic in the early days of ARPAnet

Packet Switching: Statistical Multiplexing

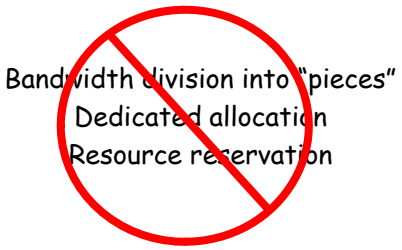


- Sequence of A & B packets does not have fixed pattern
  bandwidth shared on demand → *statistical multiplexing*
- queueing delay, packet loss

Network Core: Packet Switching

each end-end data stream divided into packets

- packets of different users *share* network resources
- each packet uses full link bandwidth

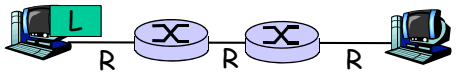


resource contention:

- aggregate resource demand can exceed amount available

  congestion: packets queue, wait for link use
- store and forward: packets move one hop at a time
  - Each node receives the ***complete*** packet before forwarding it

Disadvantage of store-and-forward

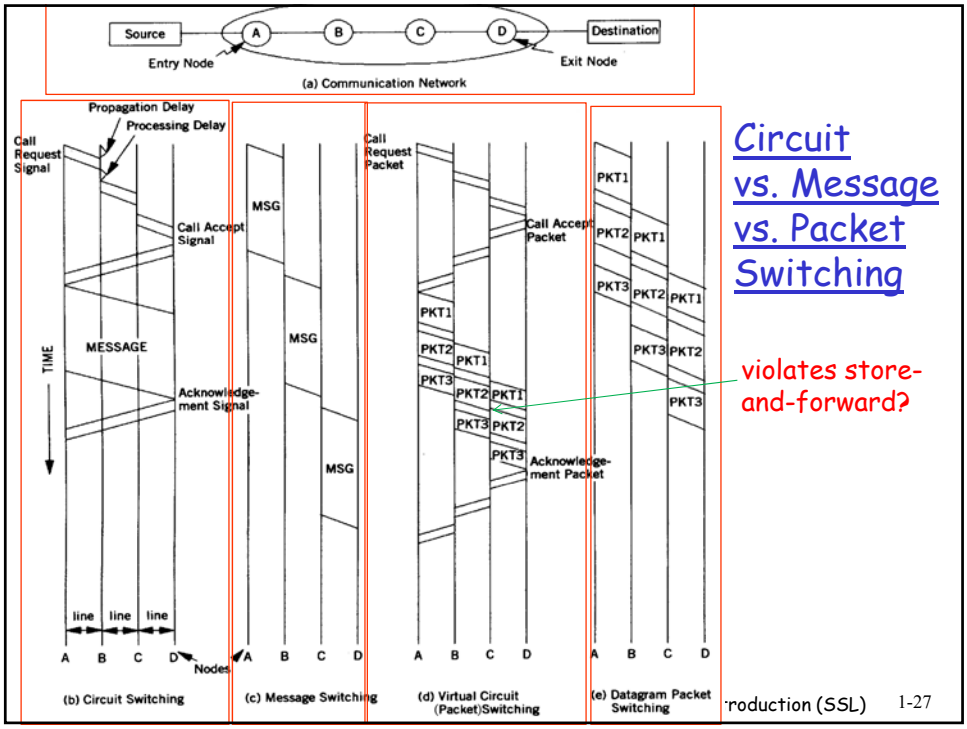


- takes L/R seconds to transmit (push out) a message of L bits into a link at R bps
- *store and forward:* entire message must arrive at router before it can be transmitted on next link

Example:

- L = 7.5 Mbits
- R = 1.5 Mbps
- End-to-end delay more than 15 seconds

- Solution: **A file/message larger than packet size is transmitted as multiple packets**



Circuit vs. Message vs. Packet Switching

violates store-and-forward?

Packet Switching versus Message Switching

Advantages of packet switching

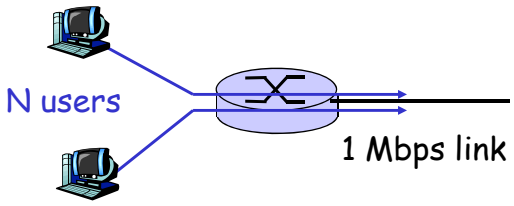
- Smaller end-to-end delay from pipelining
- Less data loss from transmission errors

Disadvantages of packet switching

- More header bits
- Additional work to do segmentation and reassembly

Packet switching versus circuit switching

- 1 Mb/s link
- each user:
  - 100 kb/s when "active"
  - active 10% of time (a "bursty" user)
- *circuit-switching:*
  - 10 users
- *packet switching:*
  - with 35 users, probability > 10 active at same time is less than .0004



Q: how did we get value 0.0004?

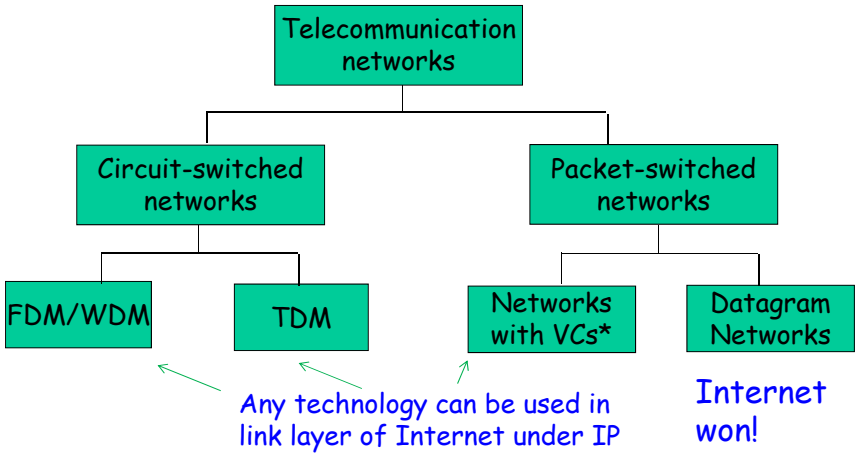
Packet switching versus circuit switching

Is packet switching a "slam dunk winner?"

- great for bursty data
  - resource sharing
  - simpler, no call setup
- excessive congestion -> packet delay and loss
  - protocols needed for reliable data transfer, congestion control
- Q: How to provide circuit-like behavior?

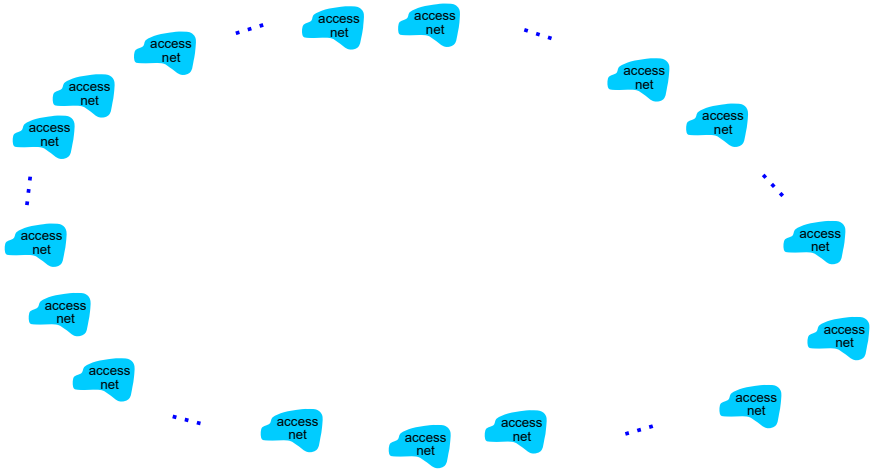
  bandwidth guarantees needed for
  - *interactive* audio/video apps
  - providing virtual links to enterprise network customers (under service contracts)

Network Taxonomy



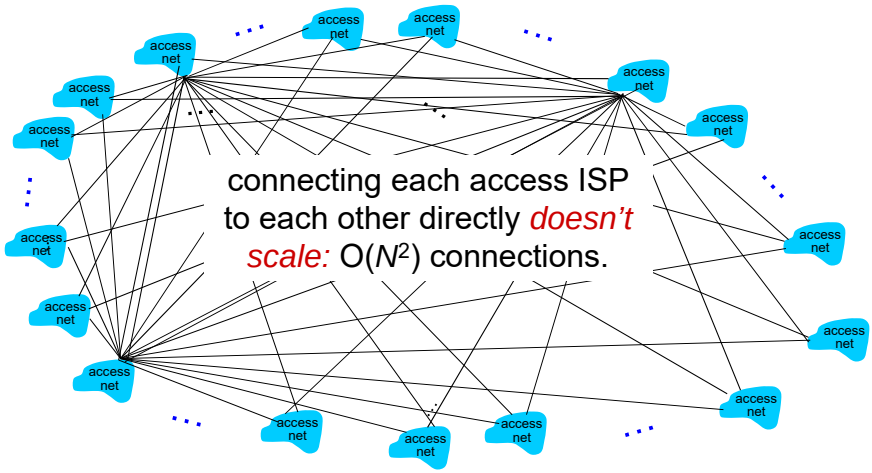
Internet structure: network of networks

Question: given *millions* of access ISPs, how to connect them together?



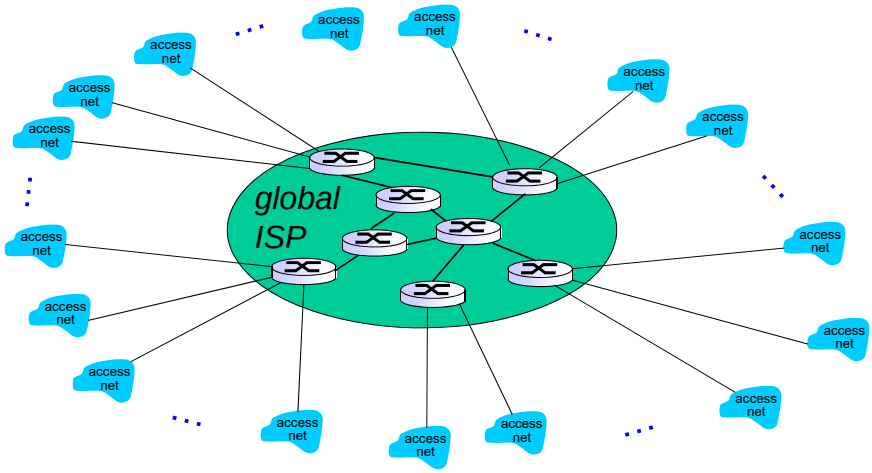
Internet structure: network of networks

Option: connect each access ISP to every other access ISP?



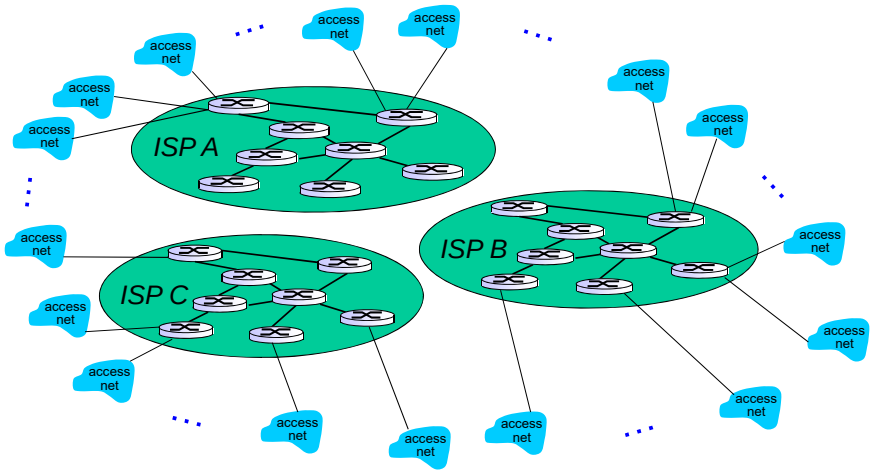
Internet structure: network of networks

Option: connect each access ISP to a global transit ISP:
1. Financed by US government: ARPAnet, NSFnet



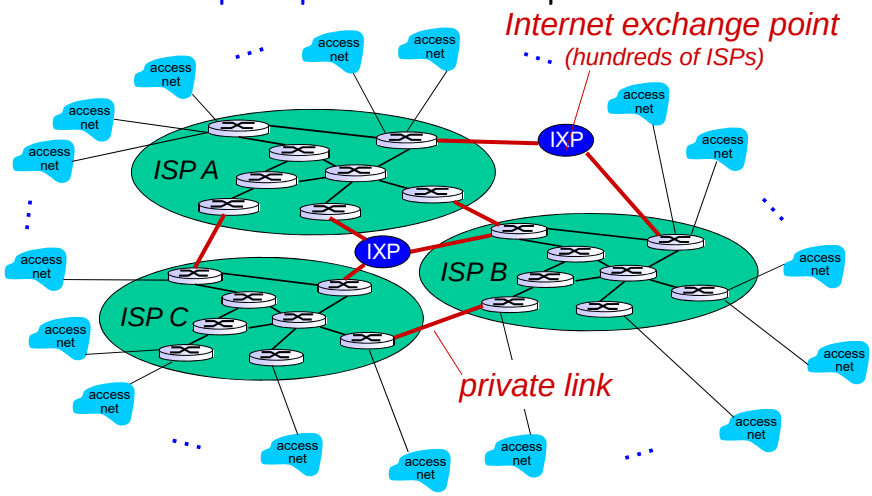
Internet structure: network of networks

But if one global ISP is viable business, there will be competitors ….



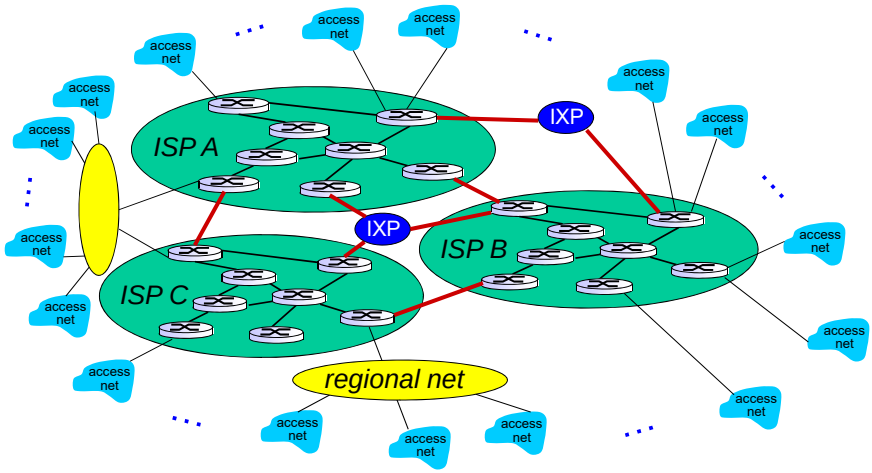
Internet structure: network of networks

But if one global ISP is viable business, there will be competitors …. Two ISPs are connected in a "provider-customer" or "peer-peer" relationship



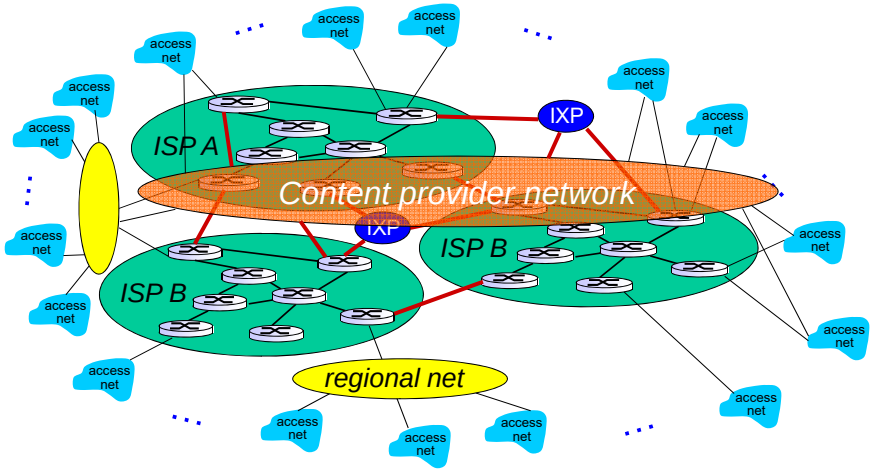
Internet structure: network of networks

… and regional networks may arise to connect access nets to ISPs

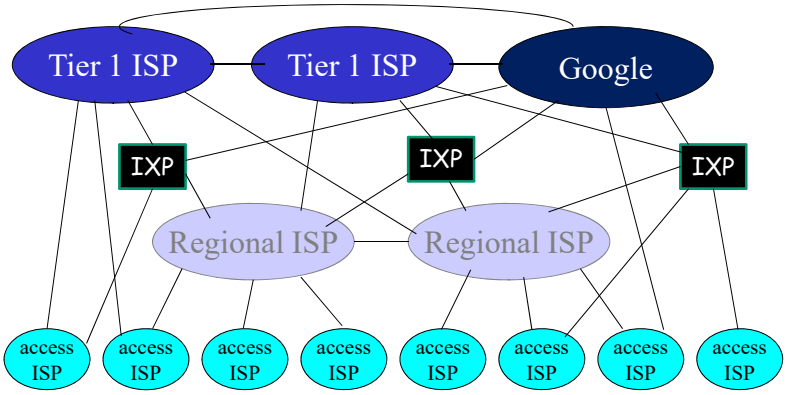


Internet structure: network of networks

… and a content provider network (e.g., Akamai, Google, Microsoft) may run its own network to bring services, content close to end users



Internet structure: network of networks



- at center: small # of well-connected large networks
  - "tier-1" commercial ISPs (e.g., Level 3, Sprint, AT&T, NTT), national & international coverage
  - content provider networks (e.g., Google): private network that connects its **data centers** to Internet, often bypassing tier-1 and regional ISPs

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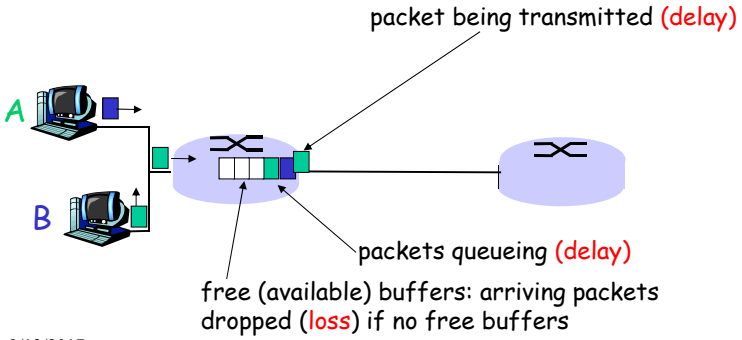
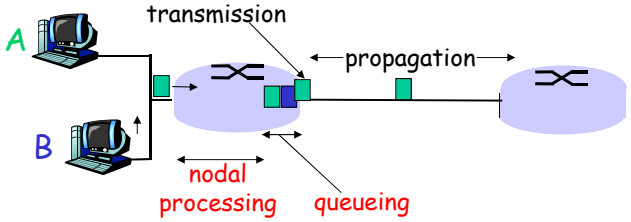
How do loss and delay occur?

- packet arrival rate to link temporarily exceeds output link capacity
- packets queue, wait for turn



Four sources of packet delay

- 1. nodal processing:
  - check bit errors
  - determine output link
- 2. queueing
  - time waiting at output link for transmission
  - depends on congestion level of router



Delay in packet-switched networks

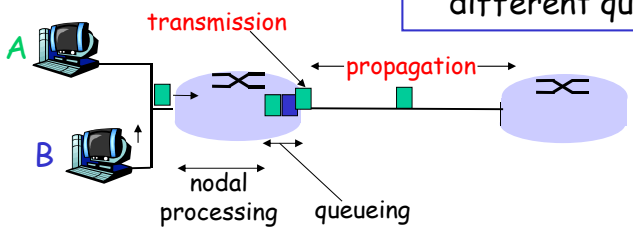
3. Transmission delay:

- R: link bandwidth (bps)
- L: packet length (bits)
- time to send bits into link = **L/R**

4. Propagation delay:

- d: length of physical link
- s: propagation speed in medium (~2×10^8 m/sec)
- propagation delay = **d/s**

Note: s and R are *very* different quantities!



End-to-End Delay

- Nodal delay (from when last bit of packet arrives at this node to when last bit arrives at next node)

  $d_{nodal} = d_{proc} + d_{queue} + d_{trans} + d_{prop}$

- End-to-end delay over N identical nodes/links from client c to server s (from when last bit of packet leaves client to when last bit arrives at server)

  $d_{c\text{-}s} = d_{prop} + Nd_{nodal}$

- Round trip time (RTT)

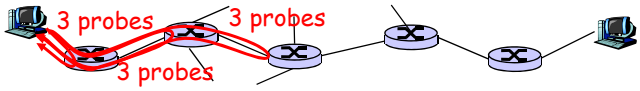
  RTT = $d_{c\text{-}s} + d_{s\text{-}c} + t_{server}$

  where t_{server} is server processing time

"Real" Internet delays and routes

- What do "real" Internet delay & loss look like?
- traceroute program: provides delay measurement from source to router along end-end Internet path towards destination.

  For all *i*:
  - sends three packets that will reach router *i* on path towards destination
  - router *i* will return packets to sender
  - sender times interval between transmission and reply.



"Real" Internet delays and routes

traceroute: gaia.cs.umass.edu to www.eurecom.fr

Three delay measurements from gaia.cs.umass.edu to cs-gw.cs.umass.edu

```
1  cs-gw (128.119.240.254)  1 ms  1 ms  2 ms
2  border1-rt-fa5-1-0.gw.umass.edu (128.119.3.145)  1 ms  1 ms  2 ms
3  cht-vbns.gw.umass.edu (128.119.3.130)  6 ms 5 ms 5 ms
4  jn1-at1-0-0-19.wor.vbns.net (204.147.132.129)  16 ms 11 ms 13 ms
5  jn1-so7-0-0-0.wae.vbns.net (204.147.136.136)  21 ms 18 ms 18 ms
6  abilene-vbns.abilene.ucaid.edu (198.32.11.9)  22 ms  18 ms  22 ms
7  nycm-wash.abilene.ucaid.edu (198.32.8.46)  22 ms  22 ms  22 ms
8  62.40.103.253 (62.40.103.253)  104 ms 109 ms 106 ms
9  de2-1.de1.de.geant.net (62.40.96.129)  109 ms 102 ms 104 ms
10  de.fr1.fr.geant.net (62.40.96.50)  113 ms 121 ms 114 ms
11  renater-gw.fr1.fr.geant.net (62.40.103.54)  112 ms  114 ms  112 ms
12  nio-n2.cssi.renater.fr (193.51.206.13)  111 ms  114 ms  116 ms
13  nice.cssi.renater.fr (195.220.98.102)  123 ms  125 ms  124 ms
14  r3t2-nice.cssi.renater.fr (195.220.98.110)  126 ms  126 ms  124 ms
15  eurecom-valbonne.r3t2.ft.net (193.48.50.54)  135 ms  128 ms  133 ms
16  194.214.211.25 (194.214.211.25)  126 ms  128 ms  126 ms
17  * * *
18  * * *
19  fantasia.eurecom.fr (193.55.113.142)  132 ms  128 ms  136 ms
```

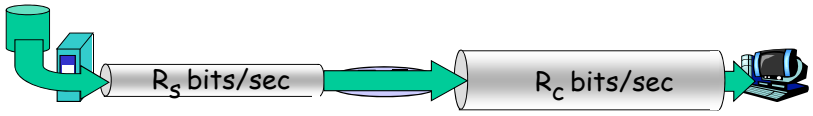
trans-oceanic link

different packets

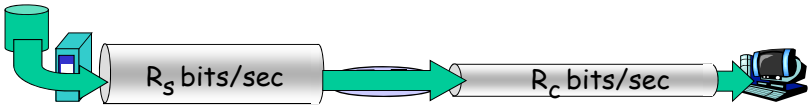
* means no response (probe lost, router not replying)

Throughput - rate at which bits are transferred from source to destination (in bits/sec.)

- $R_s < R_c$ end-end throughput less than ____ ?



- $R_s > R_c$ end-end throughput less than ____ ?



bottleneck link

link on end-end path that constrains end-end throughput

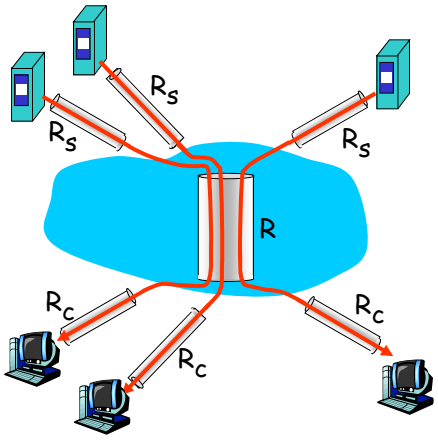
Throughput: Internet scenario

- per-connection end-to-end throughput is *approximately*

  min(R_c, R_s, R/10)

  - *Actually sharing a bottleneck equally is ideal but unrealistic*

- In practice, R_c or R_s is often the bottleneck
- or the server is the bottleneck

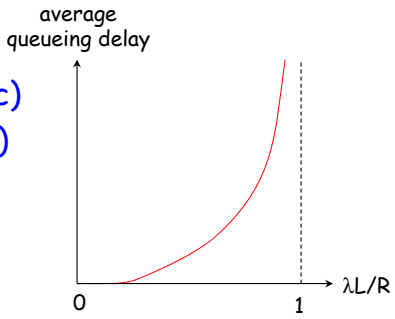


10 connections (fairly) share backbone bottleneck link R bits/sec

Queueing delay (waiting time)

- R: link bandwidth (bps)
- L: packet length (bits)
  - service rate = R/L (pkts/sec)
  - ave. service time = L/R (sec)
- λ: packet arrival rate

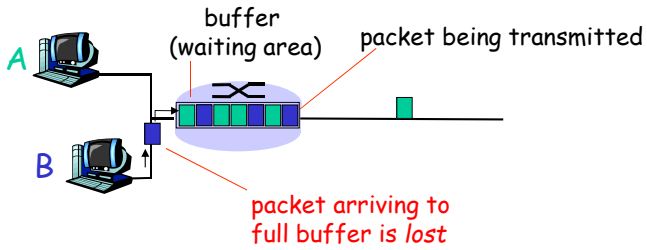
traffic intensity =
arrival rate/service rate = λL/R



- λL/R ~ 0: average queueing delay small
- λL/R -> 1: delays become large
- λL/R > 1: more "work" arriving than can be served, average delay infinite!
  - In reality, buffer overflow when λL/R -> 1

Packet loss

- buffer in router for each link has finite capacity
- lost packet may be retransmitted by previous node, by source end system, or not at all



Little's law and a useful queueing delay formula

- motivation for packet switching to replace circuit switching

Little's Law

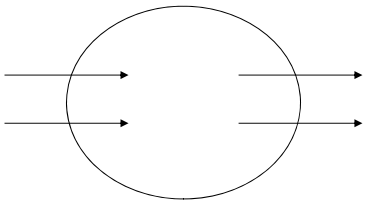
Average population = (average delay) x (throughput rate)

$$\text{average delay} = \frac{1}{N} \sum_{i=1}^{N} \text{delay}_i$$

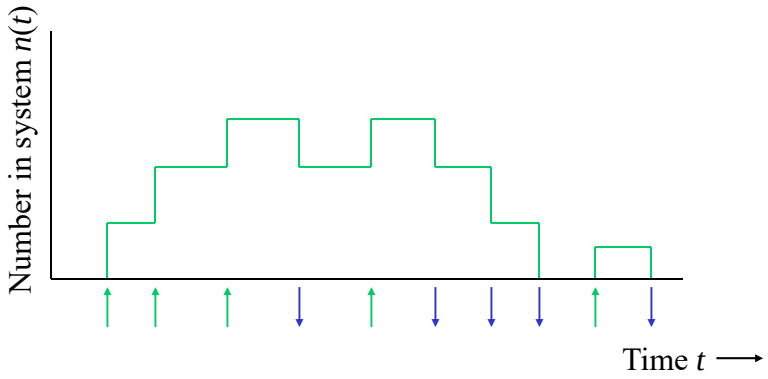
where N is number of departures

$$\text{throughput rate} = N/T$$

where T is duration of experiment



average population (to be defined)



$$\text{average population} = \frac{1}{\tau} \int_0^{\tau} n(t)dt$$

where τ is duration of the experiment

random variable x

samples $x_1, x_2, \ldots, x_n$

mean (average) $\overline{x} = \frac{1}{n} \sum_{i=1}^{n} x_i$

second moment $\overline{x^2} = \frac{1}{n} \sum_{i=1}^{n} (x_i)^2 \geq (\overline{x})^2$

$$\text{mean residual life} = \frac{\overline{x^2}}{2\overline{x}} \geq \frac{\overline{x}}{2}$$

Special case: x is a constant

$$\overline{x^2} = (\overline{x})^2$$

$$\text{mean residual life} = \frac{(\overline{x})^2}{2\overline{x}} = \frac{\overline{x}}{2}$$

random variable x

with discrete values $x_1, x_2, \ldots, x_m$

let p_i = probability $[x = x_i]$ for $i = 1, 2, \ldots, m$

by definition

mean

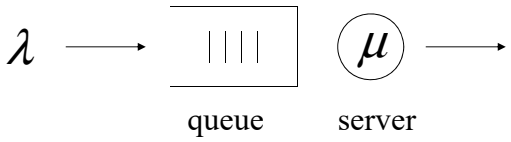
$$\bar{x} = \sum_{i=1}^{m} x_i \, p_i$$

second moment

$$\overline{x^2} = \sum_{i=1}^{m} x_i^2 \, p_i$$

(Aside: For a continuous random variable, use integration instead of summation.)

Single-Server Queue



$\bar{x}$	average service time, in seconds
μ	service rate, in packets/second ($\mu = 1/\bar{x}$)
λ	arrival rate, in packets/second
ρ	utilization of server

Conservation of flow (assuming no loss)

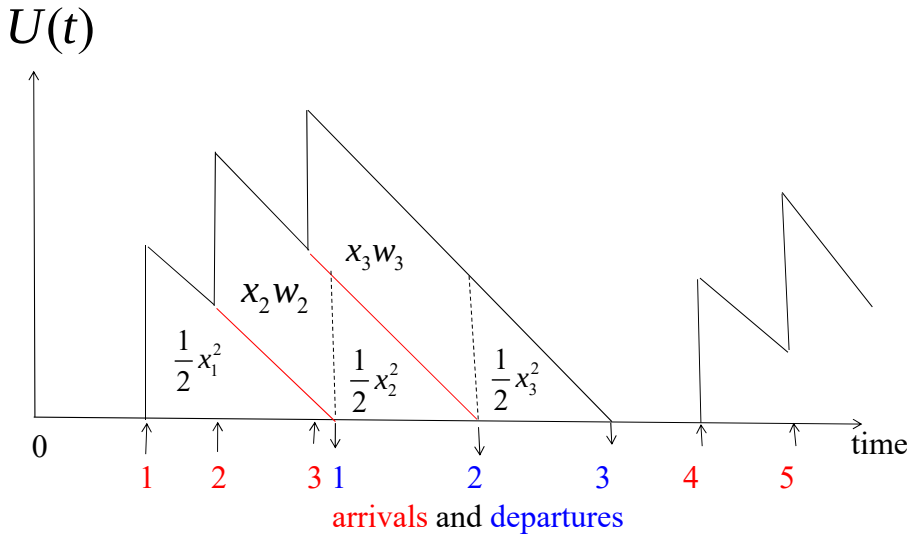
$$\lambda = \rho\mu$$

$$\rho = \frac{\lambda}{\mu} = \lambda\bar{x}$$

M/G/1 queue

- Single server
  - does not idle when there is work, no overhead, i.e., it performs 1 second of work per second
  - FIFO service
- Arrivals according to a Poisson process at rate λ packets/second
- Service times of arrivals are $x_1, x_2, \ldots, x_i \ldots$ which are *independent, identically distributed* (with a general distribution)
- Average service time is $\bar{x}$, average wait is W, average delay is T = W + $\bar{x}$

Let $U(t)$ be the unfinished work at time t



Derivation of W

Time average of unfinished work over duration τ

$$\overline{U} = \frac{1}{\tau}\int_0^{\tau} U(t)dt$$

$$= \frac{1}{\tau}\left(\frac{1}{2}\sum_{i=1}^{n} x_i^2 + \sum_{i=1}^{n} x_i w_i\right)$$

x_i and w_i are independent

$$= \frac{n}{\tau}\left(\frac{1}{2}\overline{x_i^2} + \overline{x_i}\times\overline{w_i}\right) \quad \text{where } \overline{x_i w_i} = \overline{x_i}\times\overline{w_i}$$

For Poisson arrivals, the average wait is equal to $\overline{U}$ from the *Poisson arrivals see time average (PASTA)* Theorem

Derivation of W (cont.)

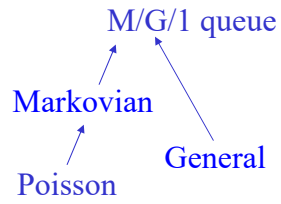
- The average wait is

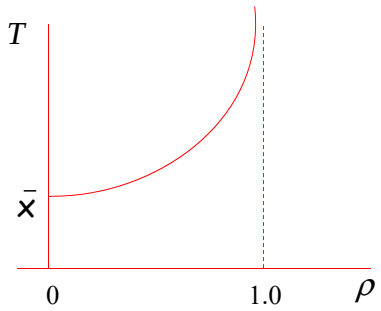
$$W = \lambda\left(\frac{1}{2}\overline{x^2} + \bar{x}W\right) = \frac{\lambda\overline{x^2}}{2} + \lambda\bar{x}W = \frac{\lambda\overline{x^2}}{2} + \rho W$$

$$W(1-\rho) = \frac{\lambda\overline{x^2}}{2}$$

$$W = \frac{\lambda\overline{x^2}}{2(1-\rho)}$$

Pollaczek-Khinchin (P-K) mean value formula





Average delay is

$$T = \bar{x} + W = \bar{x} + \frac{\lambda \overline{x^2}}{2(1-\rho)}$$

Also called Pollaczek-Khinchin (P-K) mean value formula

Special Cases

1. Service times have an exponential distribution (M/M/1). We then have

$$\overline{x^2} = 2(\bar{x})^2$$

$$W = \frac{\lambda(2)(\bar{x})^2}{2(1-\rho)} = \frac{\lambda(\bar{x})^2}{1-\rho} = \frac{\rho\bar{x}}{1-\rho}$$

$$T = W + \bar{x}$$

$$= \frac{\rho\bar{x}}{1-\rho} + \bar{x} = \frac{\rho\bar{x} + \bar{x} - \rho\bar{x}}{1-\rho}$$

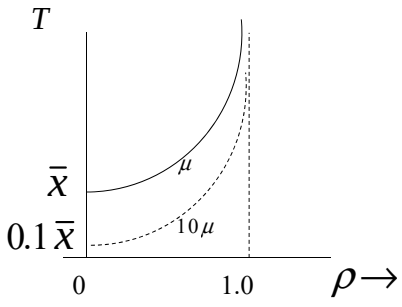
$$= \frac{\bar{x}}{1-\rho} = \boxed{\frac{\rho}{1-\rho}\frac{1}{\lambda}}$$

T decreases as λ increases

$$\lambda \to 10\lambda$$

$$\mu \to 10\mu$$

$$\boxed{\rho = \frac{10\lambda}{10\mu} = \frac{\lambda}{\mu}}$$



2. Service times are constant (deterministic)

$$\overline{x^2} = (\bar{x})^2$$

M/D/1

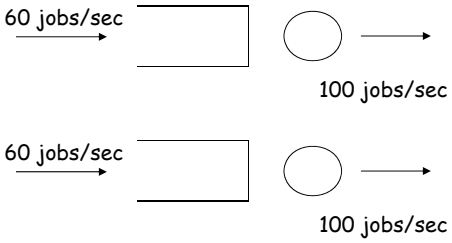
$$W = \frac{\lambda(\bar{x})^2}{2(1-\rho)} = \frac{\rho\bar{x}}{2(1-\rho)}$$

$$T = \frac{\rho\bar{x}}{2(1-\rho)} + \bar{x} = \frac{(\rho+2-2\rho)\bar{x}}{2(1-\rho)}$$

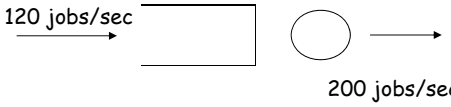
$$T = \frac{\rho(2-\rho)}{2(1-\rho)}\frac{1}{\lambda}$$

T decreases as λ increases

Two Servers and Two Queues:



Single Higher Speed Server:



Chapter 1: roadmap

1.1 What *is* the Internet?

1.2 Network edge

- end systems, access networks, links

1.3 Network core

- circuit switching, packet switching, network structure

1.4 Delay, loss and throughput in packet-switched networks

1.5 Protocol layers, service models

1.6 Networks under attack: security

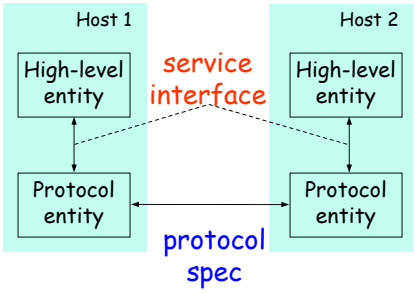
1.7 History

Layered architecture

- as **reference model** for protocol design by community effort
  - decompose a large system into smaller pieces which can be designed and implemented by different people/teams
- **modularity** eases maintenance and evolution of system
  - allows changes in implementation method so long as API remains the same, e.g., different Ethernet technologies
- strict layering **often violated** for efficient protocol implementation
  - cross-layer design

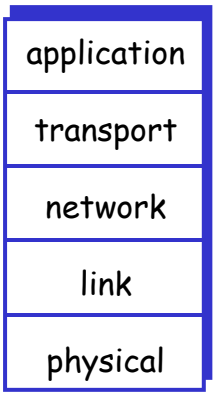
Each protocol

- involves two or more peers
- two kinds of specifications
  - service interface: operations a local user can perform on a protocol entity and get results
  - protocol spec: format and meaning of messages exchanged by protocol entities (also called peers) to provide protocol service
- The term "protocol" generally refers to protocol spec



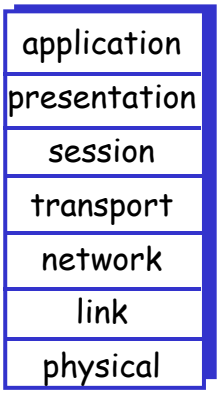
Internet protocol stack

- application: protocols that support network applications
  - SMTP, HTTP, DNS
- transport: process-process data transfer
  - TCP, UDP
- network: routing of datagrams from source to destination
  - IP, routing protocols
- link: data transfer between neighboring network elements
  - PPP, Ethernet, 802.11 (WiFi)
- physical: how to send and receive bits



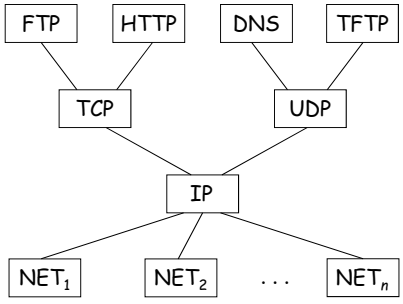
ISO/OSI reference model

- presentation: allow applications to interpret meaning of data, e.g., encryption, compression, data description (e.g. machine-specific convention)
- *session:* delimiting and synchronization of data exchange (e.g., checkpointing and recovery)
- Internet stack "missing" these layers!
  - these services, *if needed*, must be implemented in application (or application protocol)
  - *very wise decision for ARPAnet!*



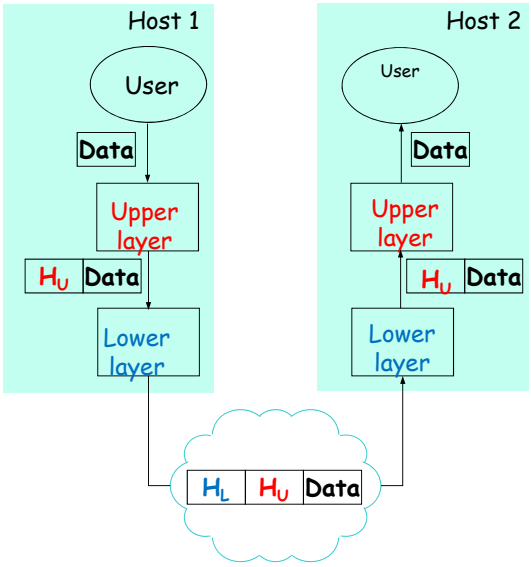
Internet Architecture

- Internet Engineering Task Force (IETF)
- application protocols support applications
- *hourglass* shape (only IP in network layer)
  - best effort service => any delivery service can be used by IP
- limitation of hourglass



Encapsulation

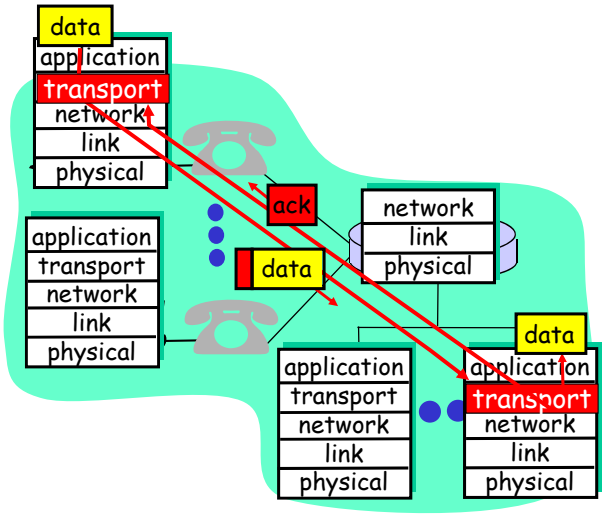
- Protocol peers provide a data delivery service
- How do protocol peers in different machines exchange protocol messages between themselves?
  - In protocol header encapsulated and de-encapsulated



Logical communication between peers

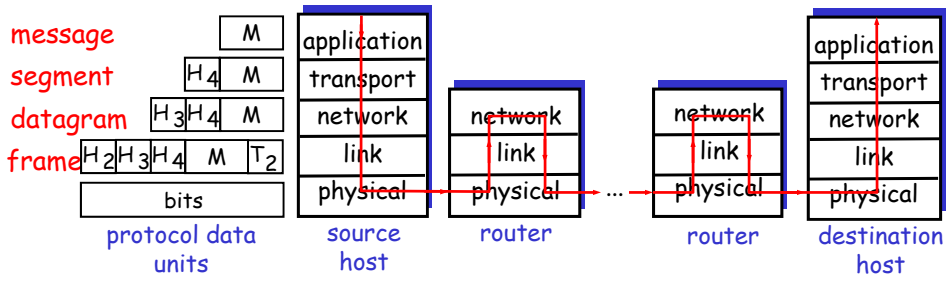
E.g.: transport

- accept data from application
- add addressing, reliability check info to form a message
- send message to peer via a delivery service
- wait for peer's reply (ack)



Physical path of data

Each layer takes data (service data unit) from above
- adds header to create its own protocol data unit
- passes protocol data unit to layer below



Note: In the past, a switch implements only two layers (physical and link). Nowadays many switches function as routers (3 layers)

Note: layer 2½ may exist (i.e. virtual circuits)

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