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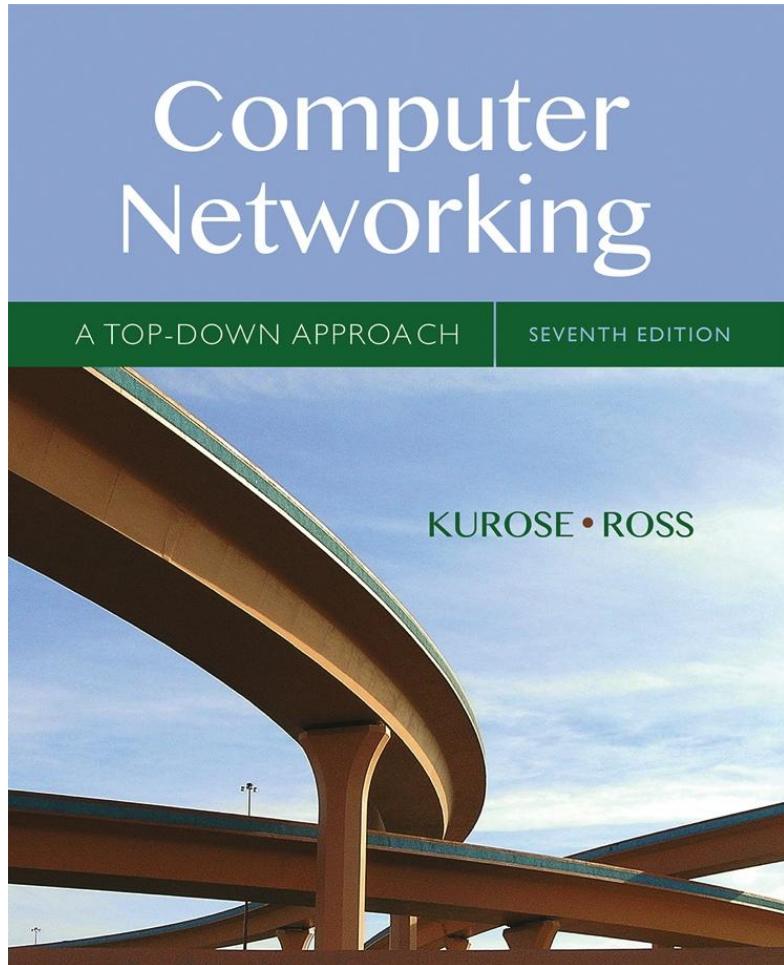
COMPUTER NETWORKS

Computer Networks and the Internet

Team Networks

Department of Computer Science and Engineering

Text Book



Computer Networking: A Top-Down Approach
Jim Kurose, Keith Ross
Pearson, 2017
7th Edition

COMPUTER NETWORKS

Computer Networks and the Internet

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Unit – 1 Computer Networks and the Internet

1.1 Introduction to Computer Networks

1.2 What is the Internet?

- A nuts-and-bolts and Services description, Protocol

1.3 Network edge

- End systems, Access networks, Physical media

1.4 Network core

- Packet switching, Circuit switching, Network structure

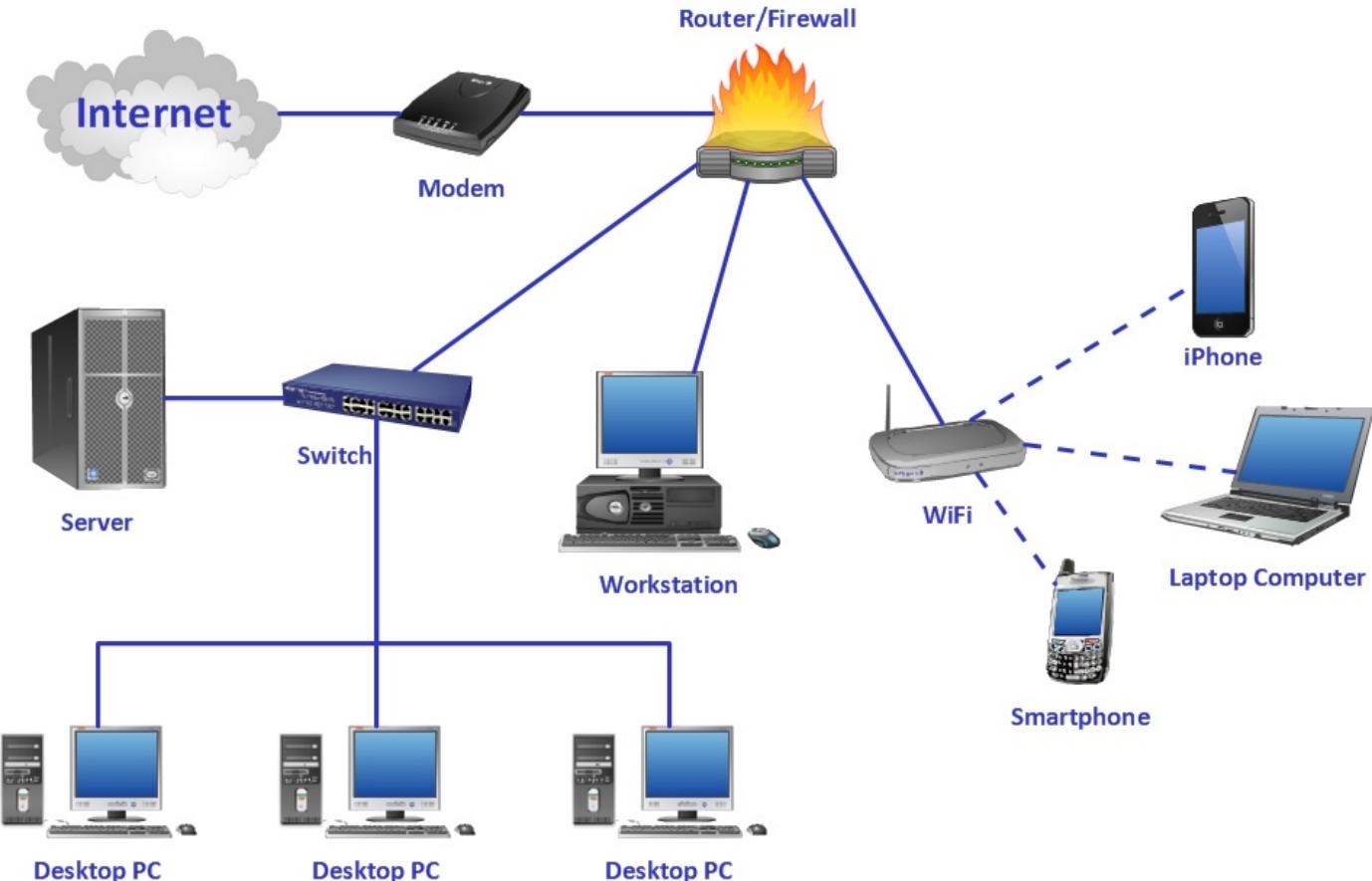
1.5 Delay, Loss & Throughput in networks

1.6 Protocol layers, Service models

- OSI model and TCP/IP protocol suite

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Introduction to Computer Networks



- Two or more devices connected together.
- Communicate with each other, share data or resources

What is the Internet?

- A massive network of networks.
- A computer network that interconnects billions of computing devices throughout the world.
- Traditional devices – PCs, Workstations, Servers – web pages, emails, etc.
- Internet “things” – laptops, PDAs, TVs, gaming consoles, home security systems, home appliances, watches, cars, traffic control systems, etc.,

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The Internet: A “Nuts and Bolts” View



Billions of connected computing *devices*:

- *hosts* = end systems
- running *network apps* at Internet's "edge"



Packet switches: forward packets (chunks of data)

- routers, switches



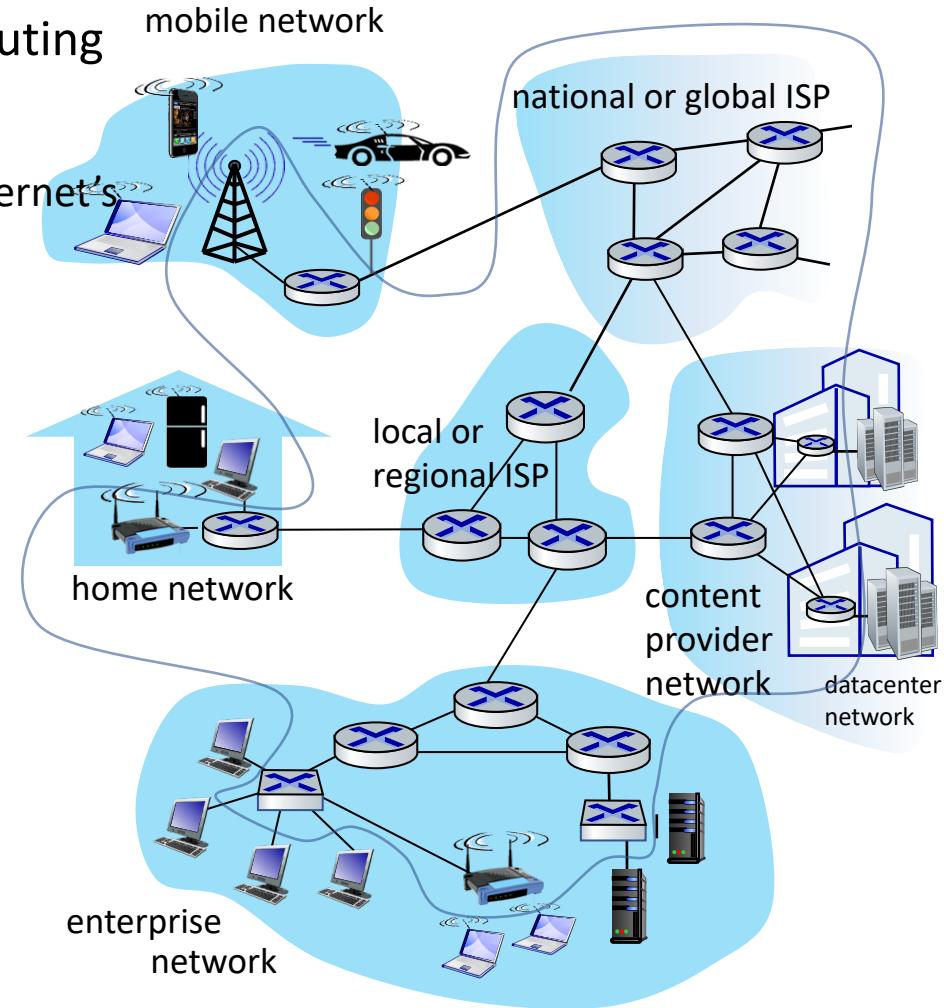
Communication links

- fiber, copper, radio, satellite
- transmission rate: *bandwidth*



Networks

- collection of devices, routers, links: managed by an organization



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“Fun” Inter-connected Devices



Amazon Echo



Internet
refrigerator



Security Camera



IP picture frame



Slingbox: remote
control cable TV



Internet phones



sensorized,
bed
mattress



Tweet-a-watt:
monitor energy
use



Web-enabled toaster +
weather forecaster



AR devices



Fitbit

Others?

**There will be 41
Billion IoT devices by
2027***

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The Internet: A “Nuts and Bolts” View

- *Internet: “network of networks”*

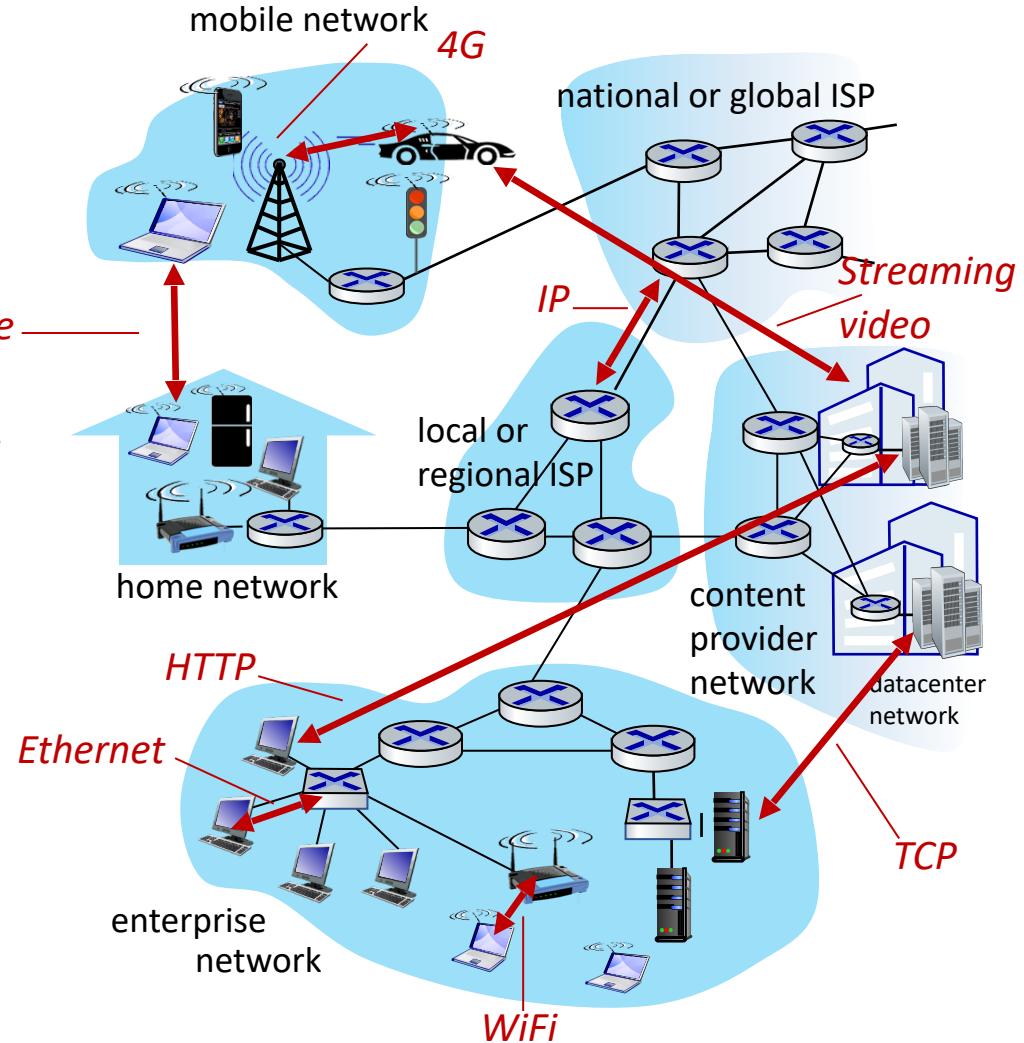
- Interconnected ISPs

- *Protocols are everywhere*

- control sending, receiving of messages
- e.g., HTTP (Web), streaming video, Skype, TCP, IP, WiFi, 4G, Ethernet

- *Internet standards*

- RFC: Request for Comments
- IETF: Internet Engineering Task Force

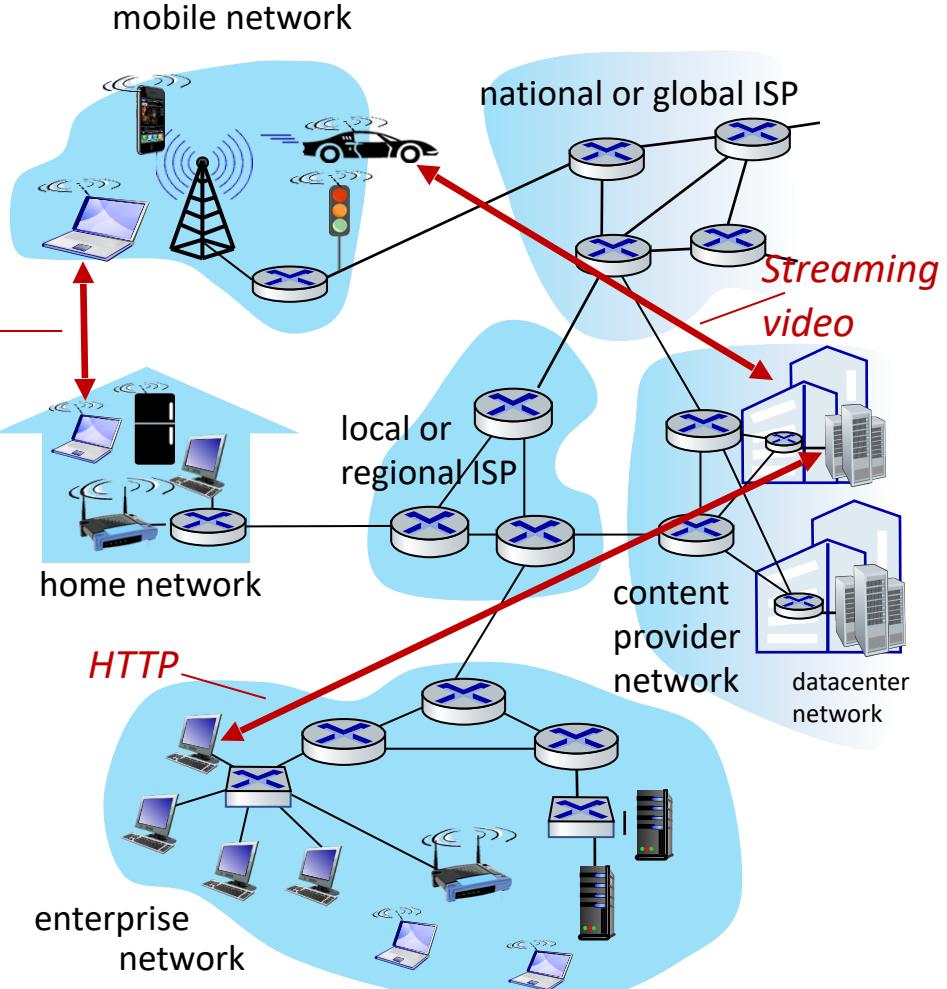


- *Infrastructure* that provides services to applications:

- Web, streaming video, multimedia teleconferencing, email, games, e-commerce, social media, inter-connected appliances, ...

- provides *programming interface* to distributed applications:

- “hooks” allowing sending/receiving apps to “connect” to, use Internet transport service
- provides service options, analogous to postal service



Human protocols:

- “what’s the time?”
- “I have a question”
- introductions

... specific messages sent

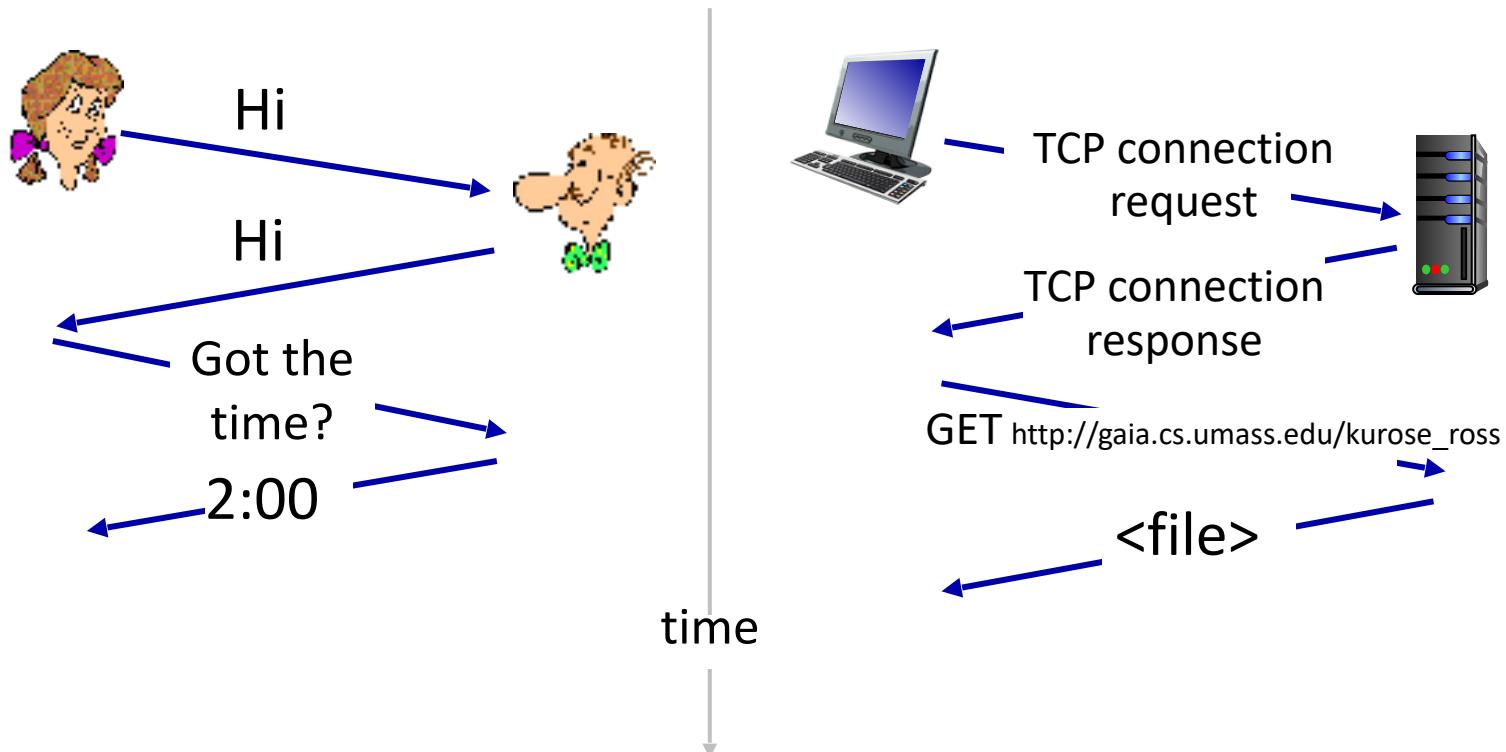
... specific actions taken when message received, or other events

Network protocols:

- computers (devices) rather than humans
- all communication activity in Internet governed by protocols

Protocols define the format, order of messages sent and received among network entities, and actions taken on msg transmission, receipt.

A human protocol and a computer network protocol:

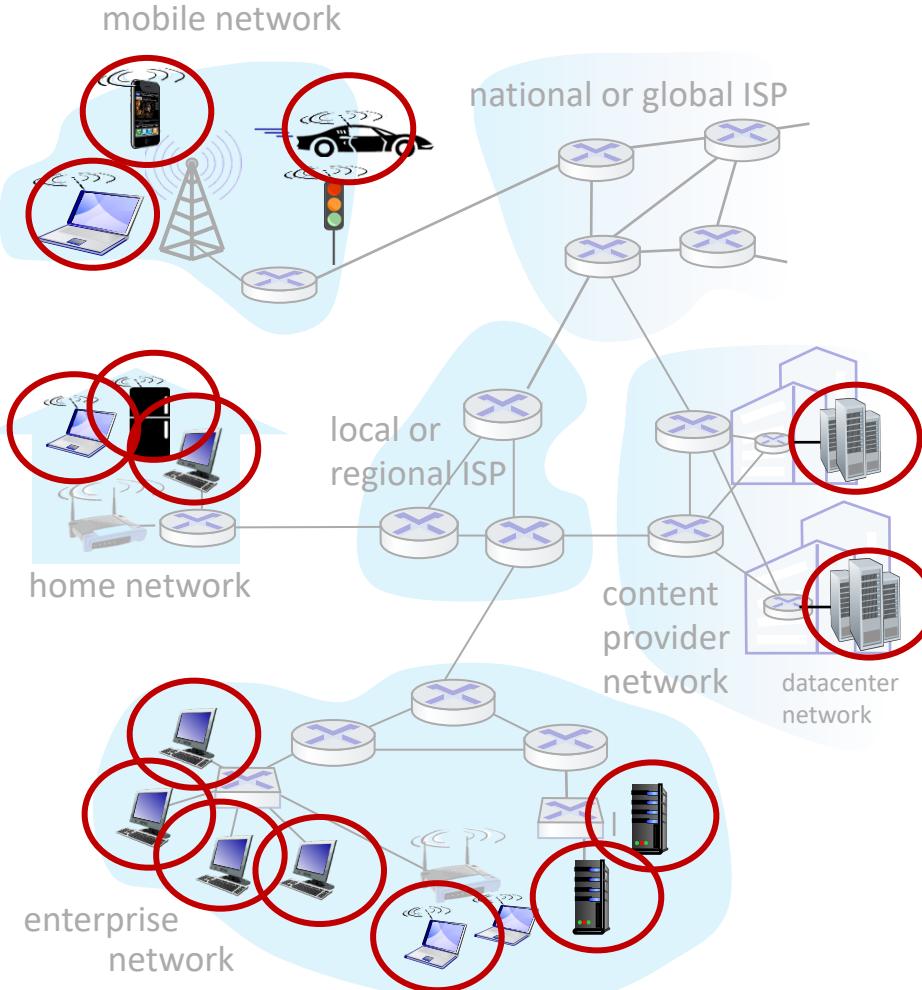


Q: other human protocols?

Network Edge: A closer look at network structure

Network edge:

- Hosts: clients & servers
- Servers in data centers



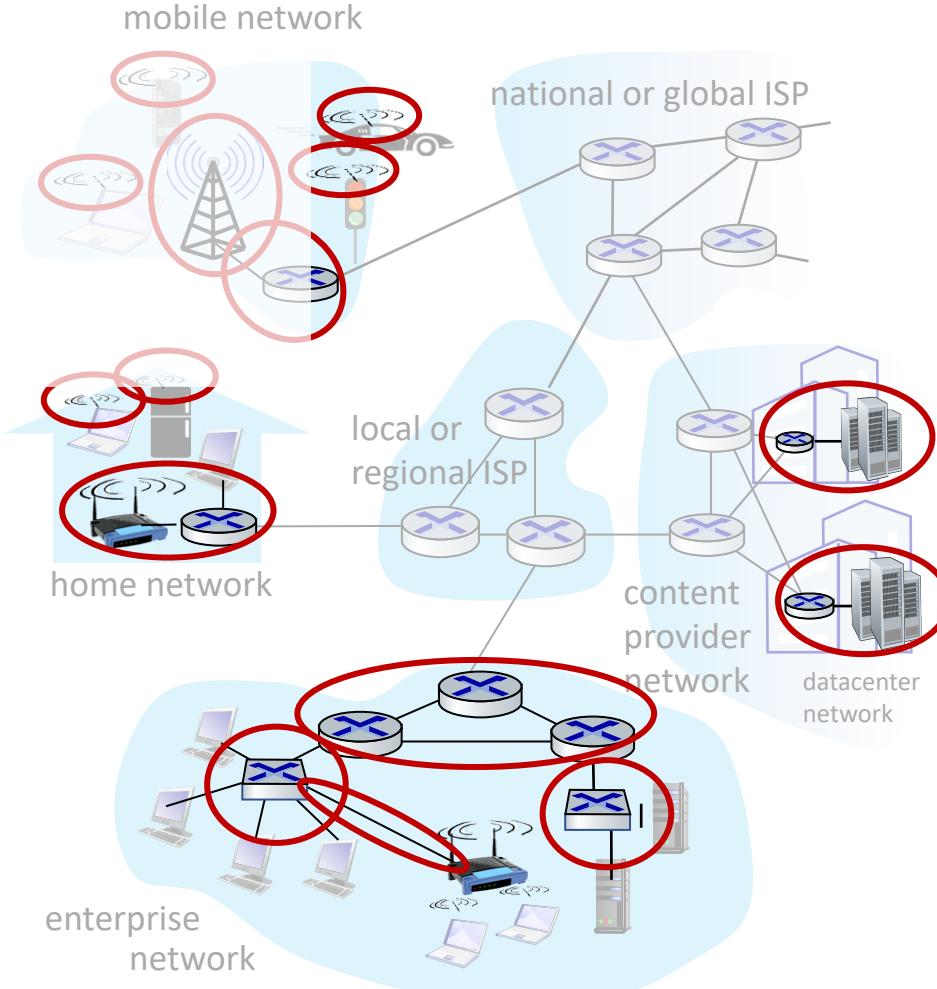
Network Edge: A closer look at network structure

Network edge:

- Hosts: clients & servers
- Servers in data centers

Access networks, physical media:

- wired, wireless communication links



Network Edge: A closer look at network structure

Network edge:

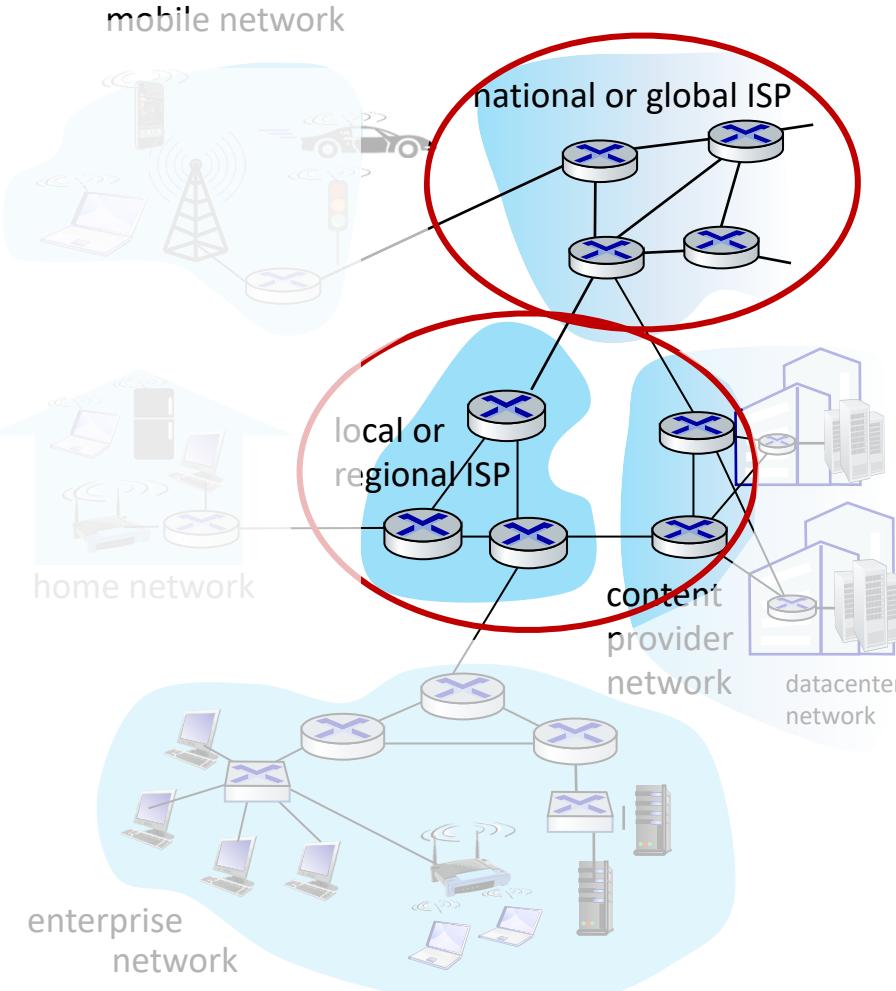
- Hosts: clients and servers
- Servers in data centers

Access networks, physical media:

- wired, wireless communication links

Network core:

- interconnected routers
- network of networks



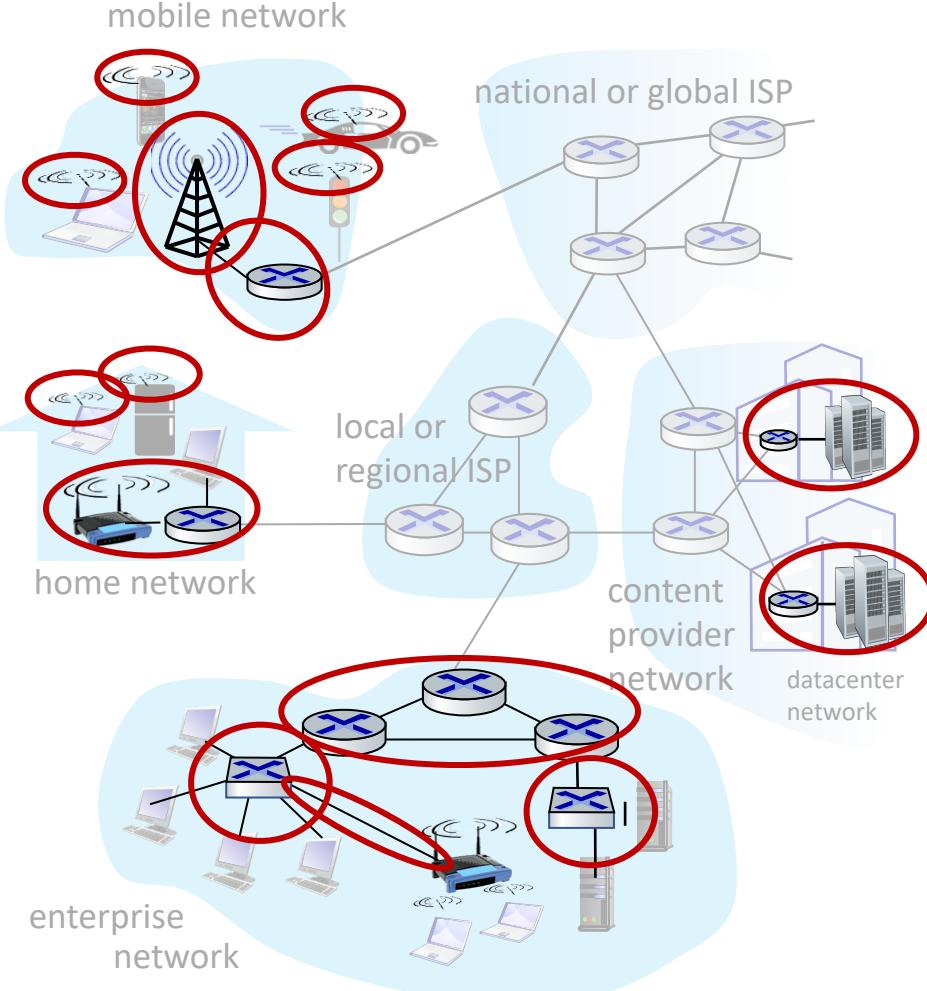
Network Edge: Access networks and Physical media

Q: How to connect end systems to edge router?

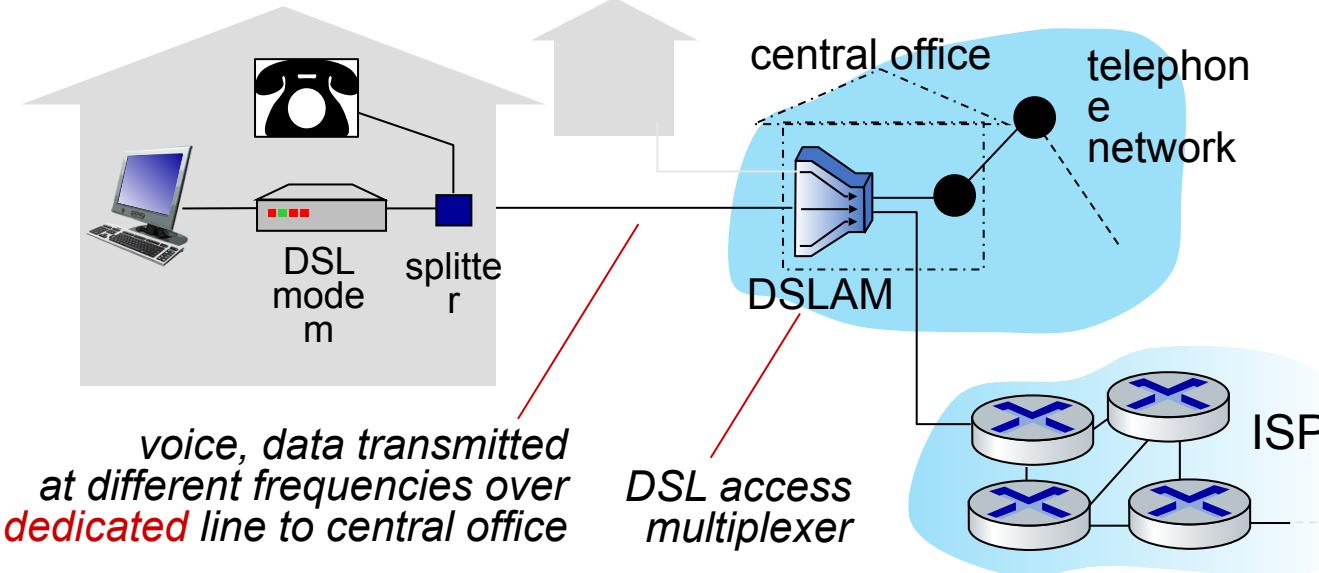
- Residential access networks
- Institutional access networks (school, company)
- Mobile access networks (WiFi, 4G/5G)

What to look for:

- Transmission rate (bits per second) of access network?
- Shared or dedicated access among users?



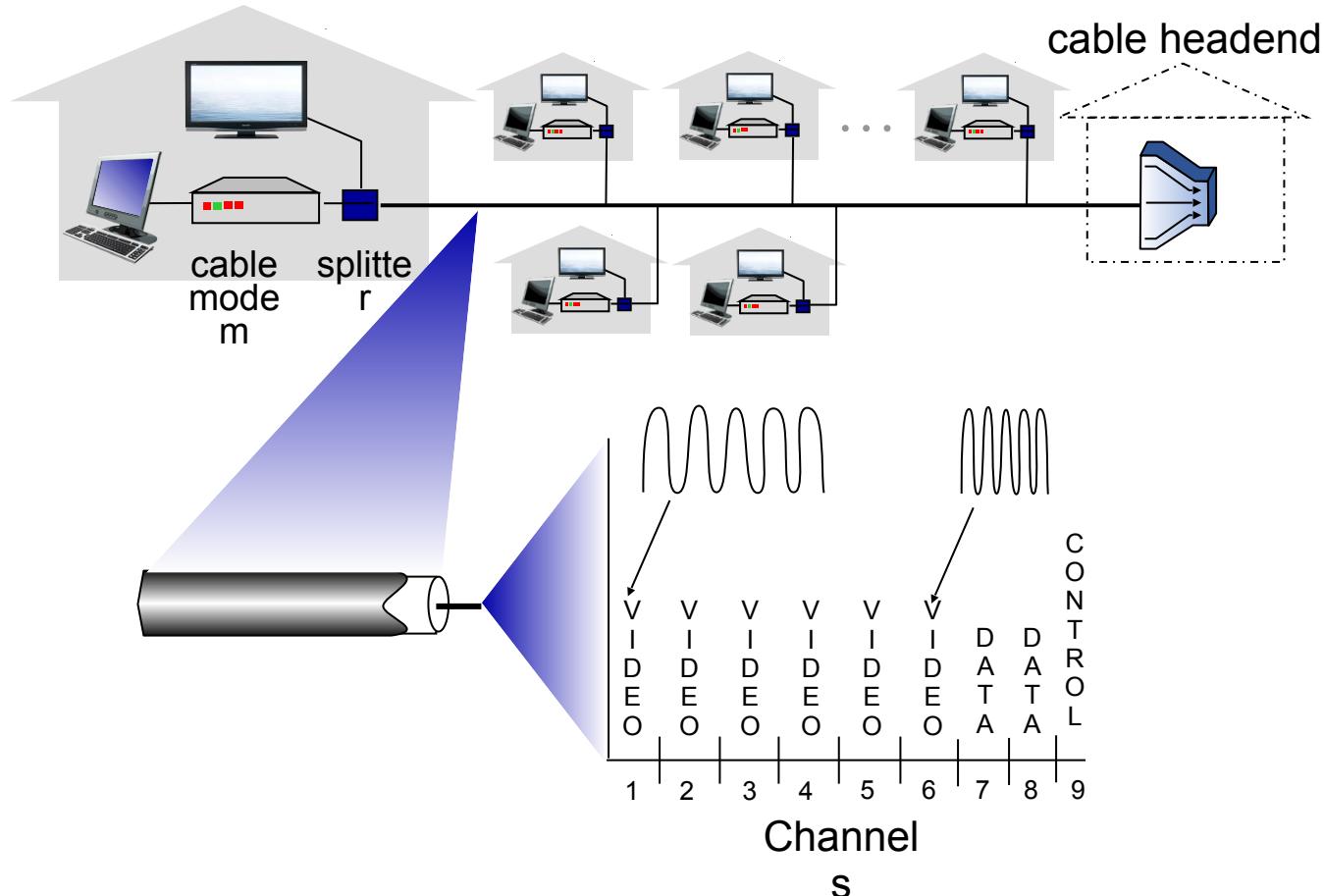
Network Edge: Access Networks - Digital Subscriber Line (DSL)



- 24-52 Mbps – downstream transmission rate
- 3.5-16 Mbps – upstream transmission rate
- Asymmetric access

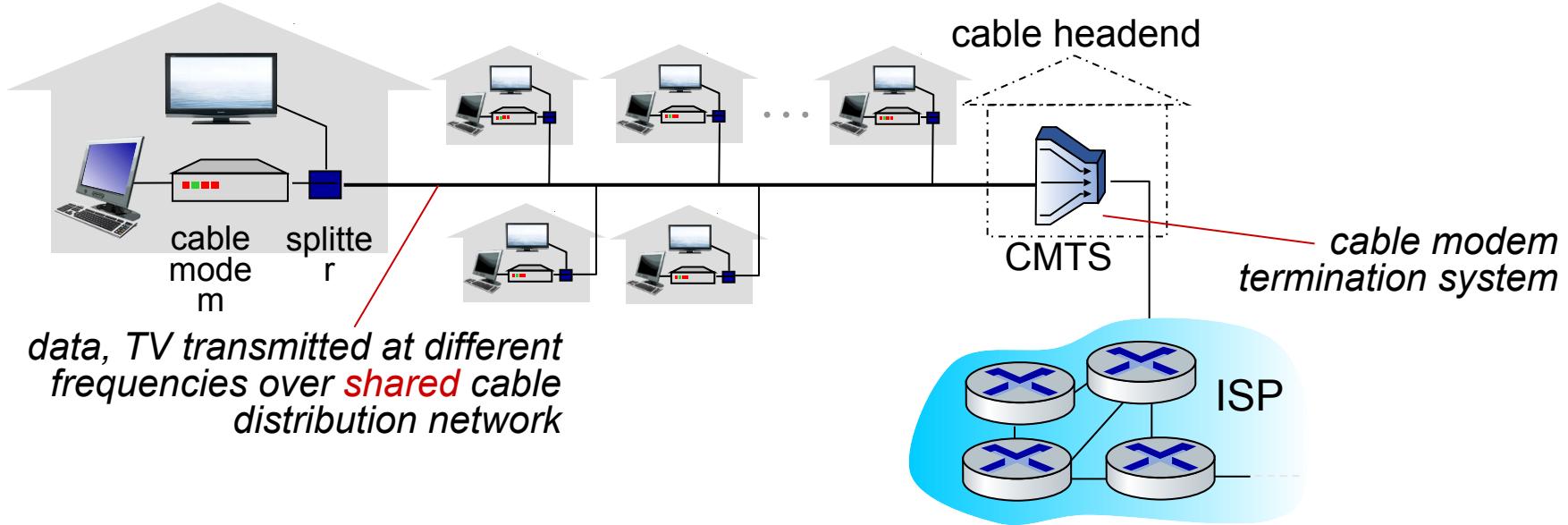
- use *existing* telephone line to central office DSLAM
 - **data** over DSL phone line goes to Internet
 - **voice** over DSL phone line goes to telephone net
- A high-speed downstream channel, in the 50 kHz to 1 MHz band
- A medium-speed upstream channel, in the 4 kHz to 50 kHz band
- An ordinary two-way telephone channel, in the 0 to 4 kHz band

Network Edge: Access Networks: Cable-based access



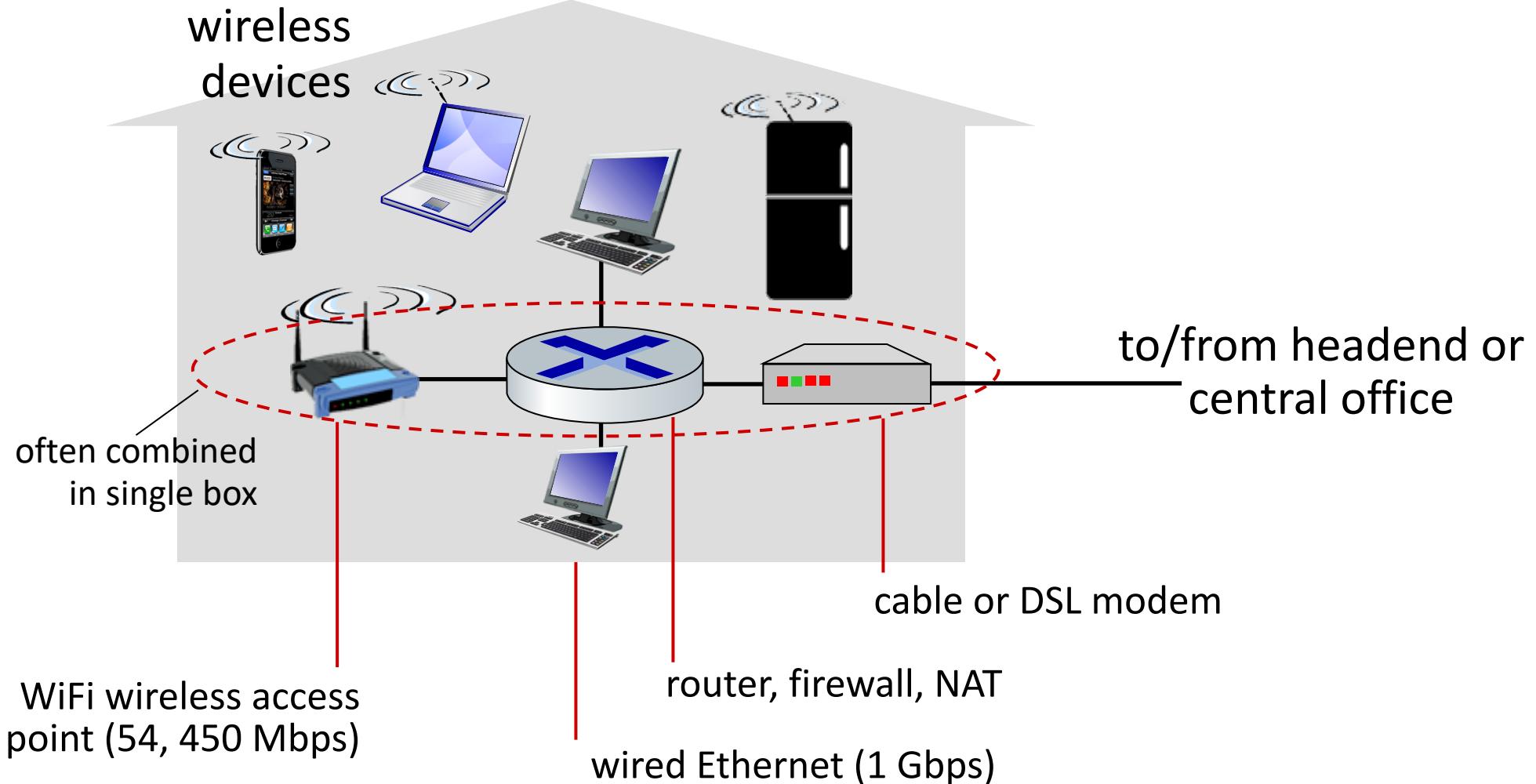
Frequency division multiplexing (FDM): different channels transmitted in different frequency bands

Network Edge: Access Networks: Cable-based access

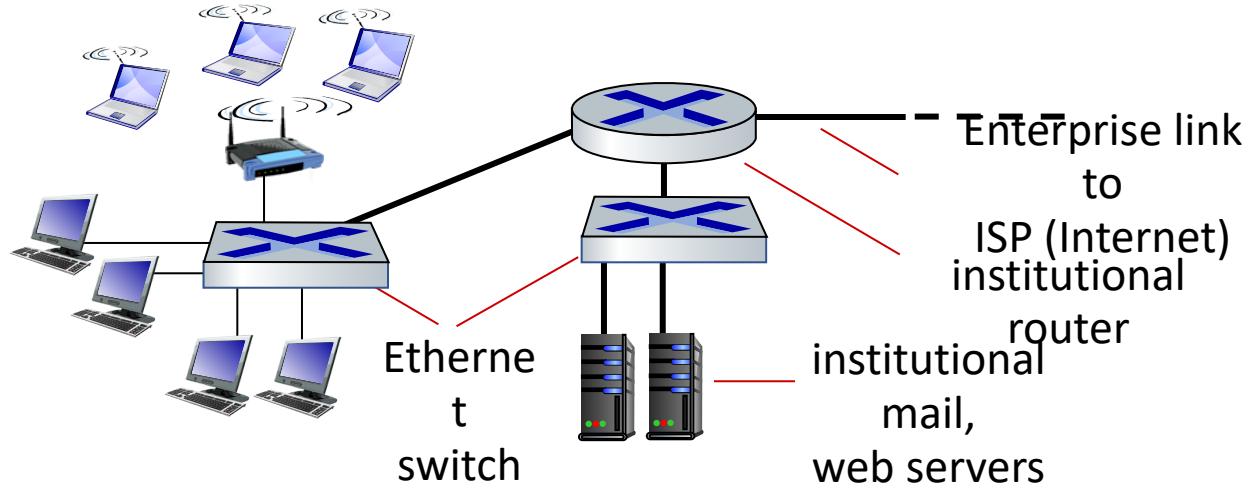


- HFC: hybrid fiber coax
 - Asymmetric:
 - up to 40 Mbps – 1.2 Gbs downstream transmission rate,
 - 30-100 Mbps upstream transmission rate
- Network of cable, fiber attaches homes to ISP router
 - homes *share access network* to cable headend

Network Edge: Access Networks – Home access



Network Edge: Access Networks – Enterprise networks



- companies, universities, etc.
- mix of wired, wireless link technologies, connecting a mix of switches and routers (we'll cover differences shortly)
 - Ethernet: wired access at 100Mbps, 1Gbps, 10Gbps
 - WiFi: wireless access points at 11, 54, 450 Mbps

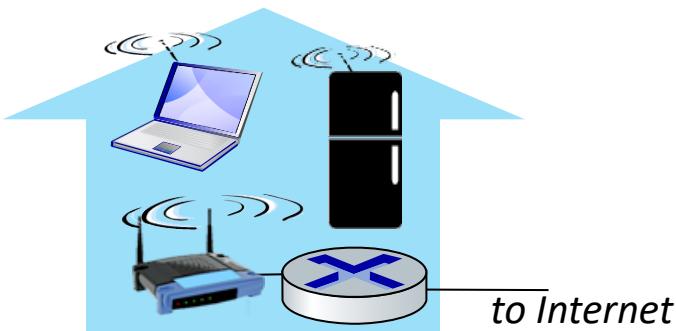
Network Edge: Wireless Access Networks

Shared *wireless* access network connects end system to router

- via base station aka “access point”

Wireless local area networks (WLANS)

- typically within or around building (~100 ft)
- 802.11b/g/n (WiFi): 11, 54, 450 Mbps transmission rate



Wide-area cellular access networks

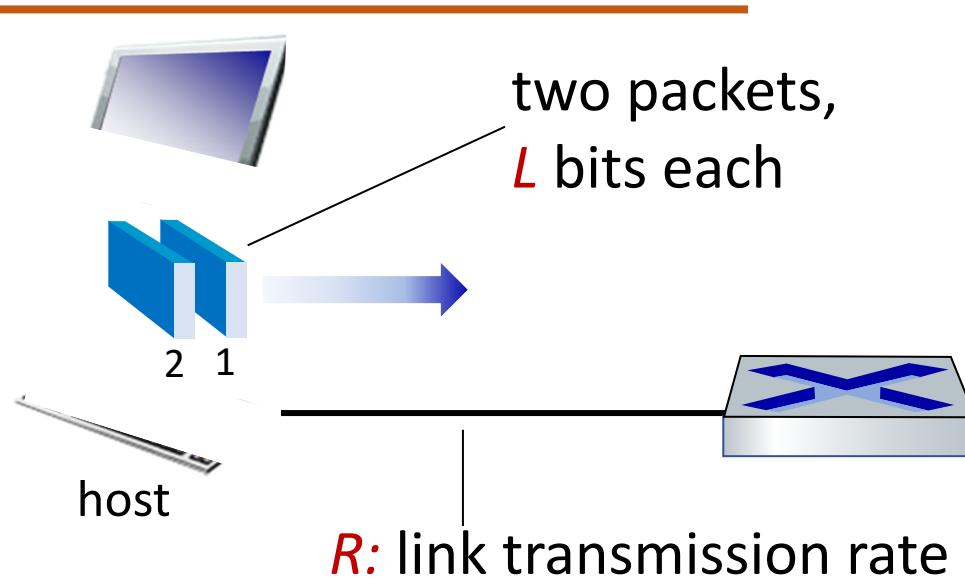
- provided by mobile, cellular network operator (10's km)
- 10's Mbps
- 4G cellular networks (5G coming)



Hosts: Send packets of data

Host sending function:

- takes application message
- breaks into smaller chunks, known as *packets*, of length L bits
- transmits packet into access network at *transmission rate R*
 - link transmission rate, aka link *capacity, aka link bandwidth*



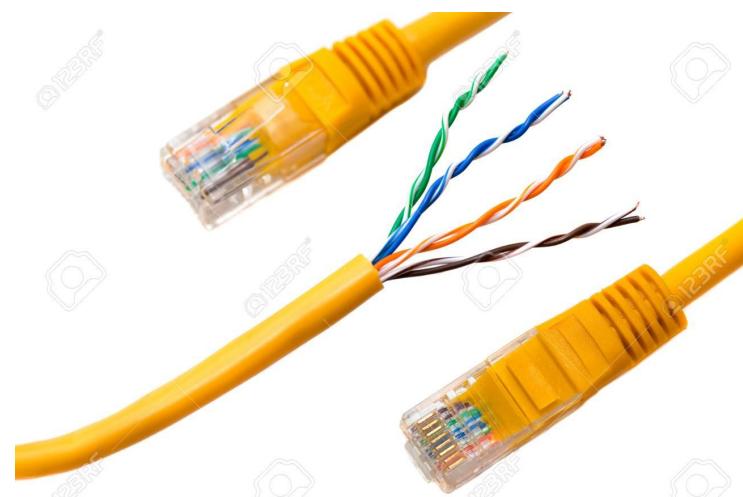
$$\text{packet transmission delay} = \frac{\text{time needed to transmit } L\text{-bit packet into link}}{\text{link transmission rate}} = \frac{L \text{ (bits)}}{R \text{ (bits/sec)}}$$

Network Edge: Physical media

- **bit:** propagates between transmitter/receiver pairs
- **physical link:** what lies between transmitter & receiver
- **guided media:**
 - signals propagate in solid media: copper, fiber, coax
- **unguided media:**
 - signals propagate freely, e.g., radio

Twisted pair (TP)

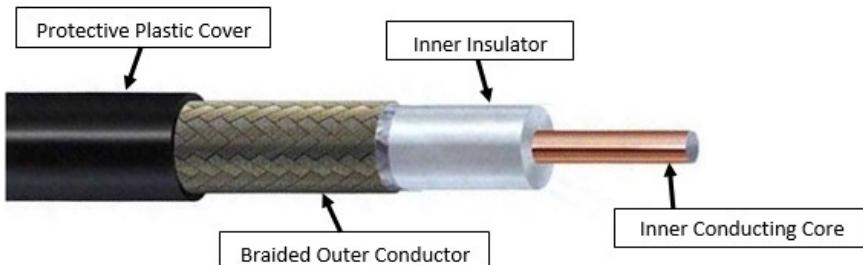
- two insulated copper wires (STP & UTP)
 - Category 5: 100 Mbps, 1 Gbps Ethernet
 - Category 6: 10Gbps Ethernet



Network Edge: Physical media

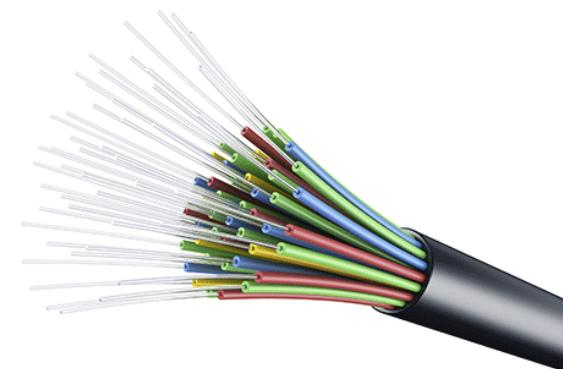
Coaxial cable:

- two concentric copper conductors
- concentric rather than parallel
- bidirectional
- broadband:
 - multiple frequency channels on cable
 - 100's Mbps per channel



Fiber optic cable:

- glass fiber carrying light pulses, each pulse a bit
- high-speed operation:
 - high-speed point-to-point transmission (10's-100's Gbps)
- low error rate:
 - repeaters spaced far apart
 - immune to electromagnetic noise



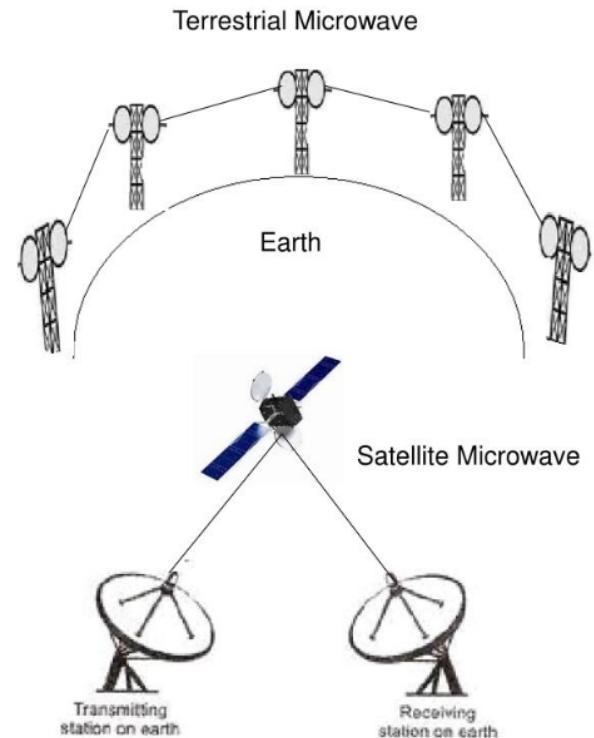
Network Edge: Physical media

Wireless radio

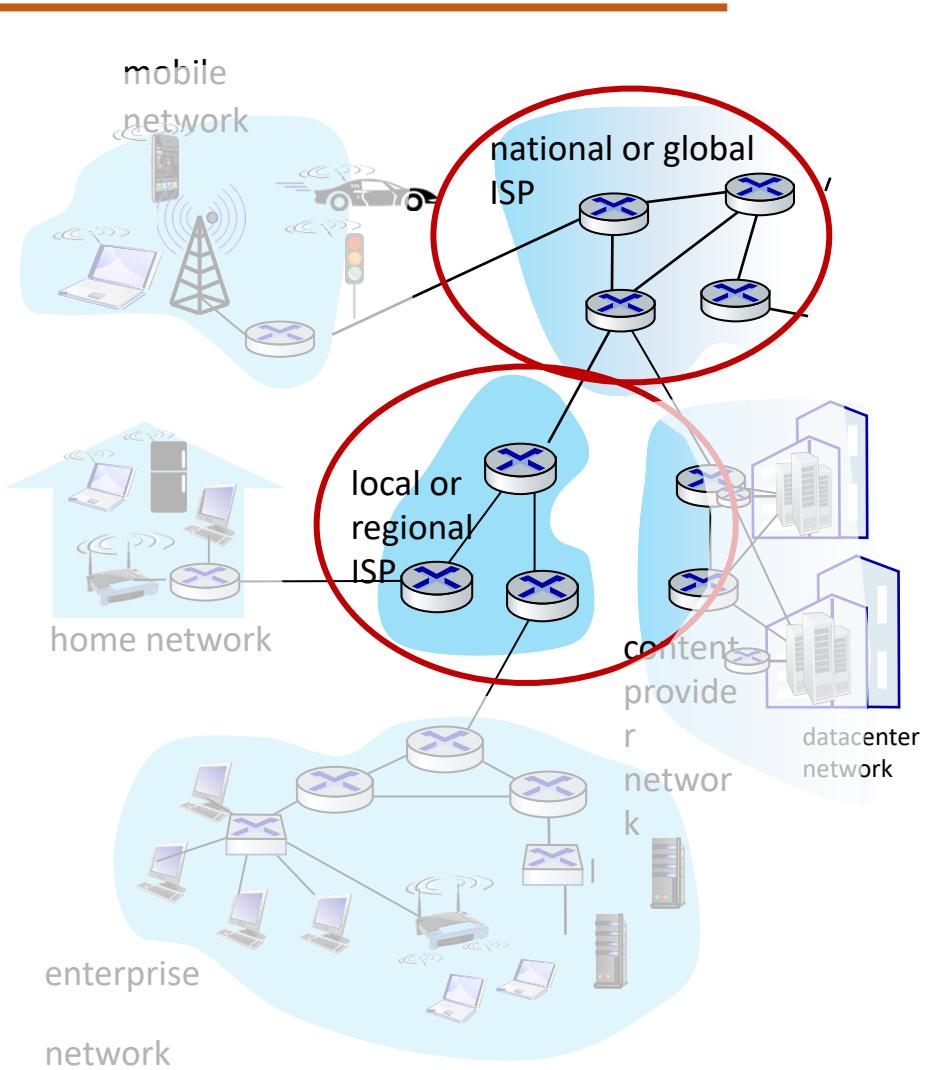
- signal carried in electromagnetic spectrum
- no physical “wire”
- broadcast and “half-duplex” (sender to receiver)
- propagation environment effects:
 - reflection
 - obstruction by objects
 - interference

Radio link types:

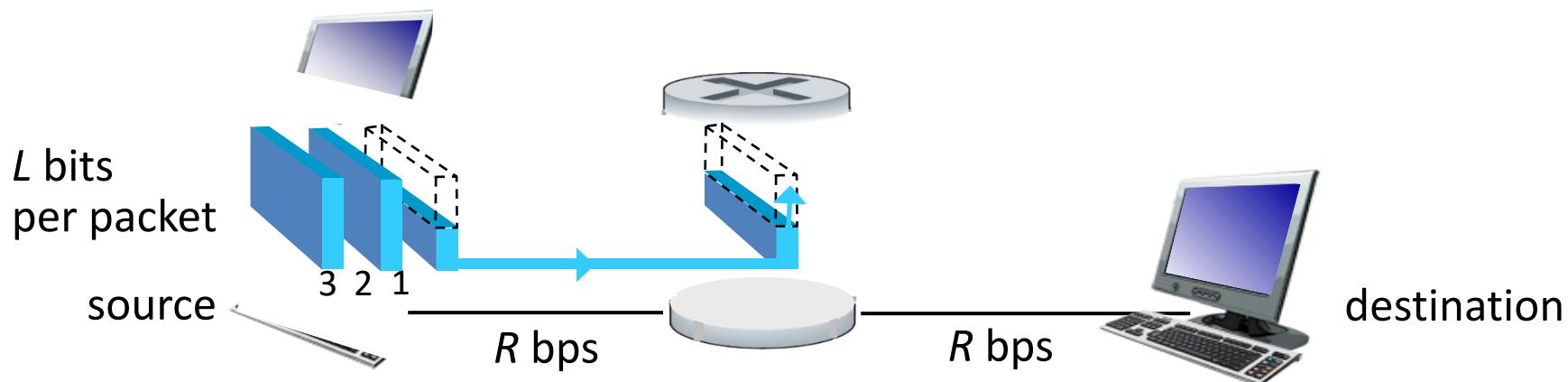
- terrestrial microwave
 - up to 45 Mbps channels
- **Wireless LAN (WiFi)**
 - Up to 100's Mbps
- **wide-area** (e.g., cellular)
 - 4G cellular: ~ 10's Mbps
- **satellite**
 - up to 45 Mbps per channel
 - 280 msec end-end delay
 - geosynchronous vs. low-earth-orbit



- mesh of interconnected routers
- **packet-switching:** hosts break application-layer messages into *packets*
 - forward packets from one router to the next, across links on path from source to destination
 - each packet transmitted at full link capacity



Network Core: Packet Switching: store-and-forward

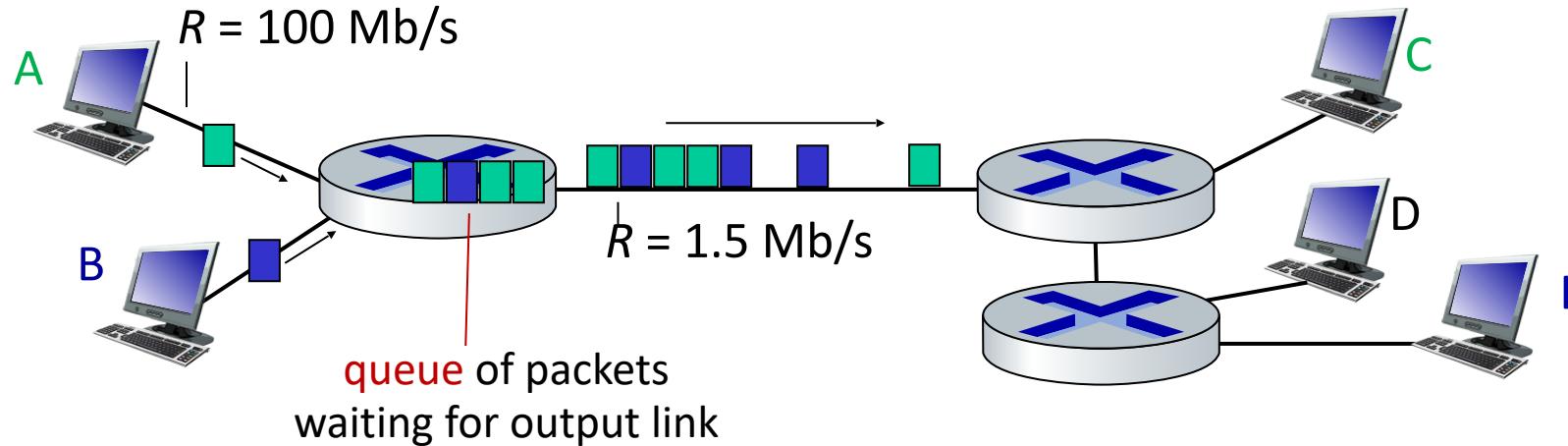


- **Transmission delay:** takes L/R seconds to transmit (push out) L -bit packet into link at R bps
- **Store and forward:** entire packet must arrive at router before it can be transmitted on next link
- **End-end delay:** $2L/R$ (above), assuming zero propagation delay (more on delay shortly)

One-hop numerical example:

- $L = 10$ Kbits
- $R = 100$ Mbps
- one-hop transmission delay = 0.1 msec

Network Core: Packet Switching: queuing delay, loss



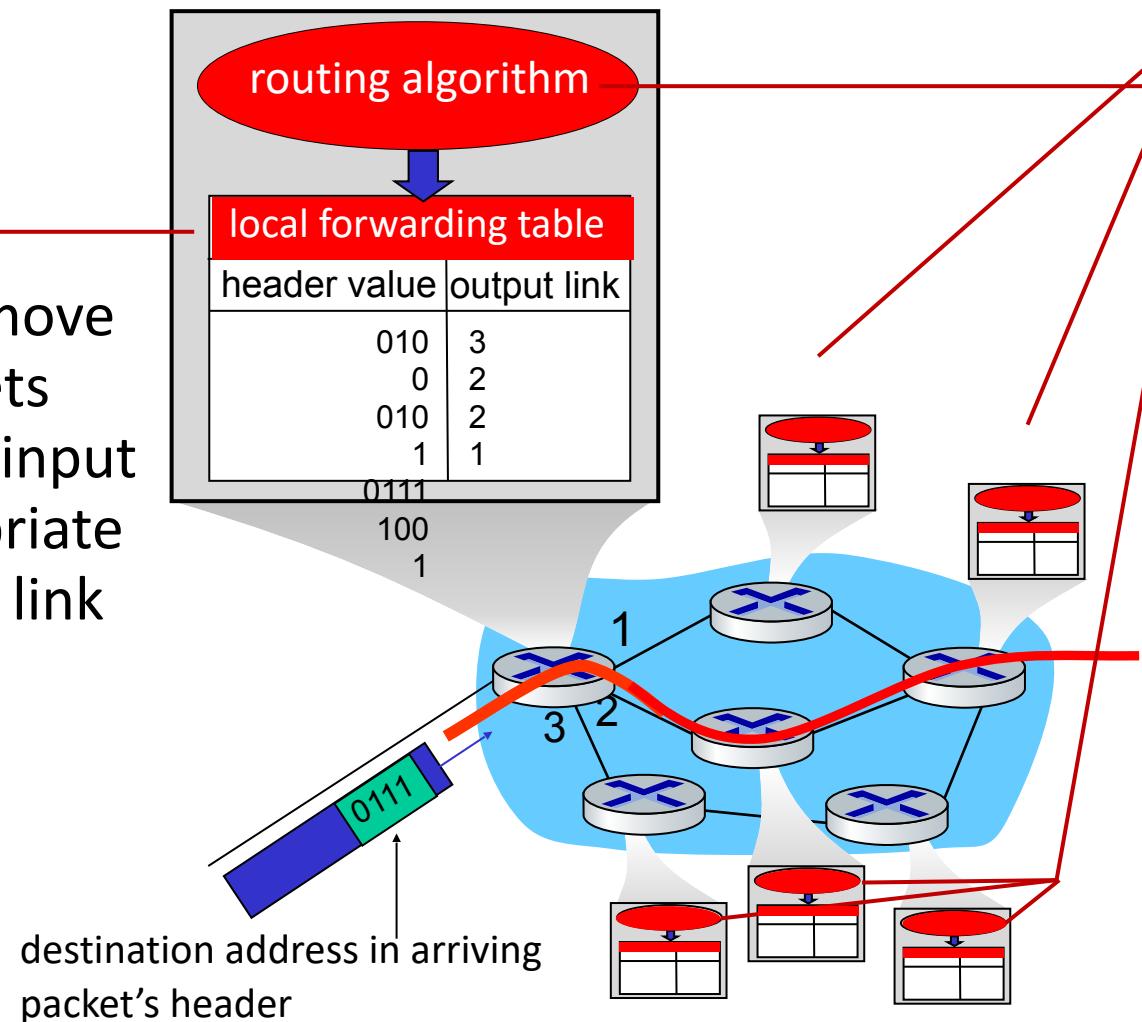
Packet queuing and loss: if arrival rate (in bps) to link exceeds transmission rate (bps) of link for a period of time:

- packets will queue, waiting to be transmitted on output link
- packets can be dropped (lost) if memory (buffer) in router fills up

Network Core: Two Key Network Core Functions

Forwarding:

- *local* action: move arriving packets from router's input link to appropriate router output link



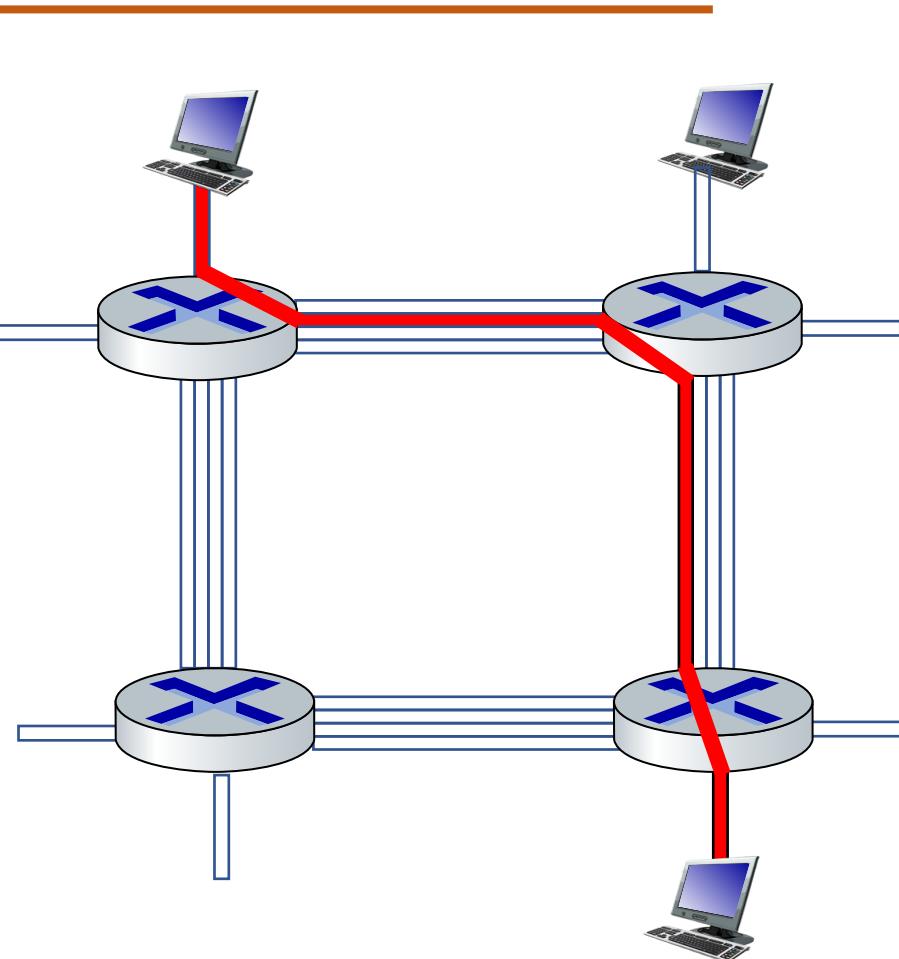
Routing:

- *global* action: determine source-destination paths taken by packets
- routing algorithms

Network Core: Circuit Switching

end-end resources allocated to, reserved for “call” between source and destination (eg: telephone)

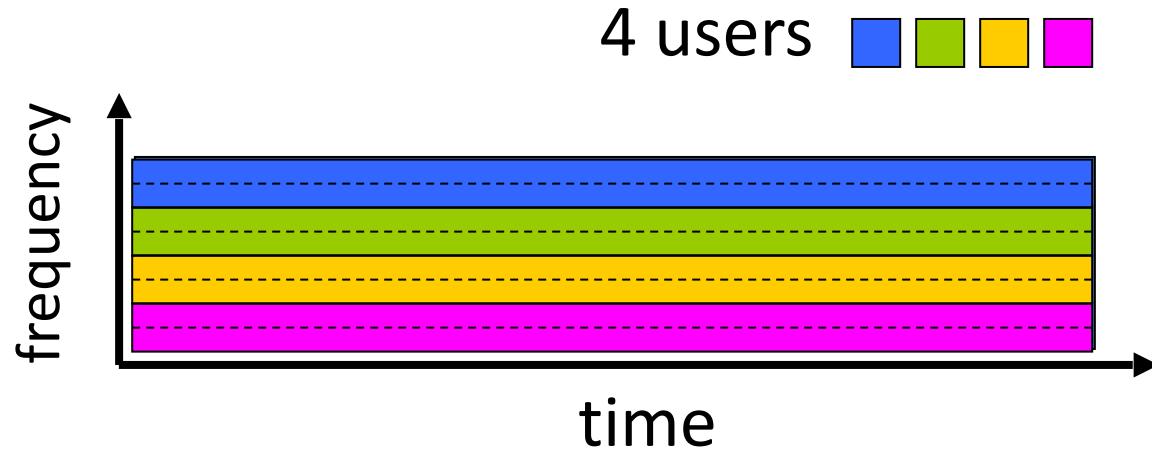
- in diagram, each link has four circuits.
 - call gets 2nd circuit in top link and 1st circuit in right link.
- dedicated resources: no sharing
 - circuit-like (guaranteed) performance
- circuit segment idle if not used by call (no sharing)
- commonly used in traditional telephone networks



Multiplexing in Circuit Switched Networks: FDM & TDM

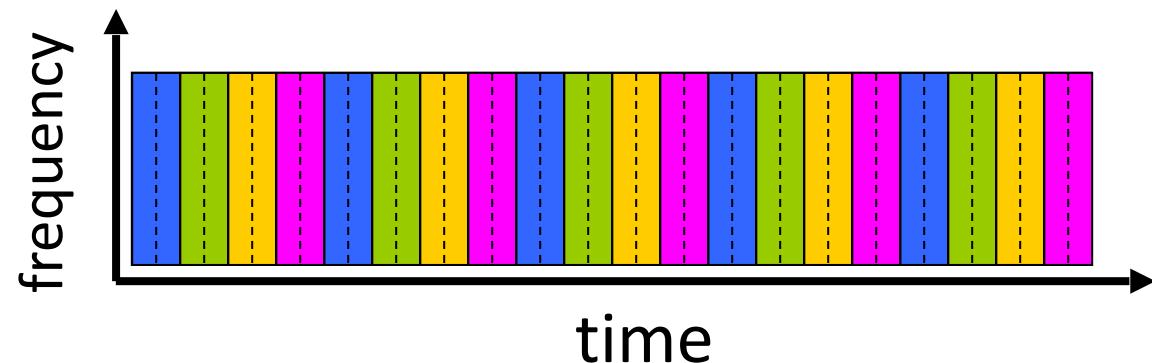
Frequency Division Multiplexing (FDM)

- optical, electromagnetic frequencies divided into (narrow) frequency bands
- each call allocated its own band, can transmit at max rate of that narrow band



Time Division Multiplexing (TDM)

- time divided into frames -> slots
- each call allocated periodic slot(s), can transmit at maximum rate of (wider) frequency band, but only during its time slot(s)

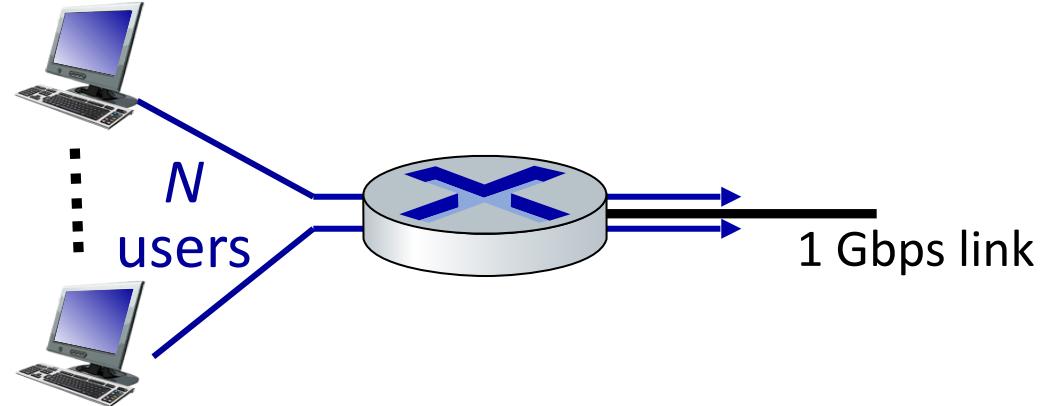


Network Core: Packet Switching vs Circuit Switching

packet switching allows more users to use network!

Example:

- 1 Gb/s link
- each user:
 - 100 Mb/s when “active”
 - active 10% of time



- *circuit-switching*: 10 users

Q: how did we get value 0.0004?

- *packet switching*: with 35 users,

Q: what happens if > 35 users ?

- probability > 10 active users at same time is less than .0004 *
- 10 or few active users, probability 0.9996

* Check out the online interactive exercises for more examples: http://gaia.cs.umass.edu/kurose_ross/interactive

Network Core: Packet Switching vs Circuit Switching

Is packet switching a “slam dunk winner”?

- great for “bursty” data – sometimes has data to send, but at other times not
 - resource sharing
 - simpler, no call setup
- **excessive congestion possible:** packet delay and loss due to buffer overflow
 - protocols needed for reliable data transfer, congestion control
- **Q: How to provide circuit-like behavior?**
 - bandwidth guarantees traditionally used for audio/video applications

Q: human analogies of reserved resources (circuit switching) versus on-demand allocation (packet switching)?

Packet Switching vs Circuit Switching – Numerical Example

- How long does it take to send a file of 640,000 bits (1 byte = 8 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
 - 500 msec to establish end-to-end circuit

Let's work it out!

Solution:

- Each circuit has a transmission rate of $(1.536 \text{ Mbps})/24 = 64 \text{ kbps}$
- It takes $(640,000 \text{ bits})/(64 \text{ kbps}) = 10 \text{ seconds}$ to transmit the file
- To this 10 seconds we add the circuit establishment time, giving 10.5 seconds to send the file



Network Core: Packet Switching vs Circuit Switching

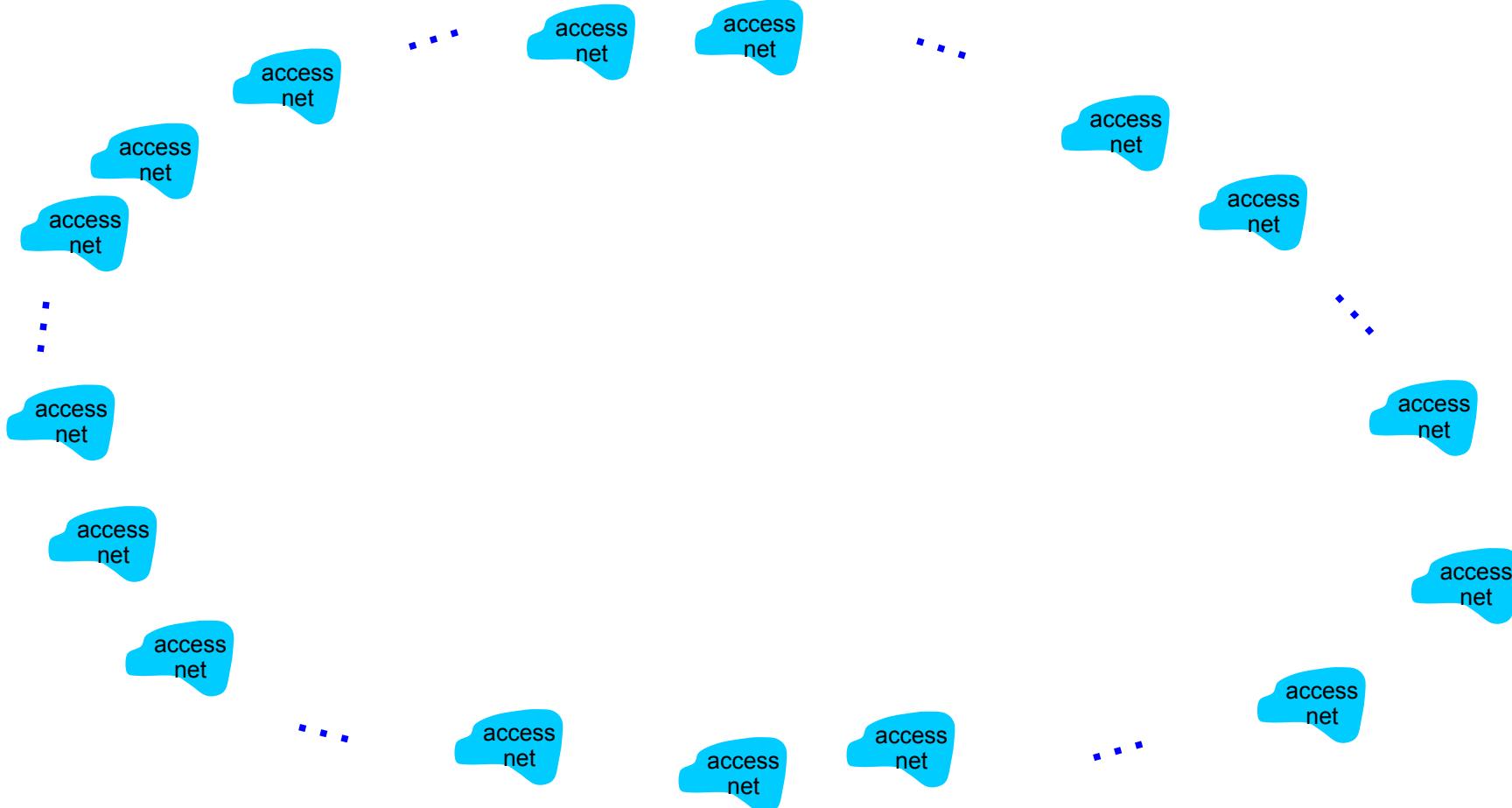
- Connectionless
 - Designed for data
 - Flexible
 - Out of order, assembled at the dest
 - Forward, Store & Fwd
 - Network layer
 - Bandwidth is saved (dynamic)
 - Transmission of data – Source, routers
 - Transmission delay
-
- Connection oriented
 - Designed for voice
 - Inflexible
 - Message received in same order
 - FDM & TDM
 - Physical layer
 - Bandwidth is wasted (fixed)
 - Transmission of data – source
 - Call setup delay

Internet Structure: a “network of networks”

- ❖ End systems connect to Internet via **access ISPs** (Internet Service Providers)
 - Residential, company and university ISPs
- ❖ Access ISPs in turn must be interconnected.
 - ❖ So that any two hosts can send packets to each other
- ❖ Resulting network of networks is very complex
 - ❖ Evolution was driven by **economics** and **national policies**
- ❖ Let's take a stepwise approach to describe current Internet structure

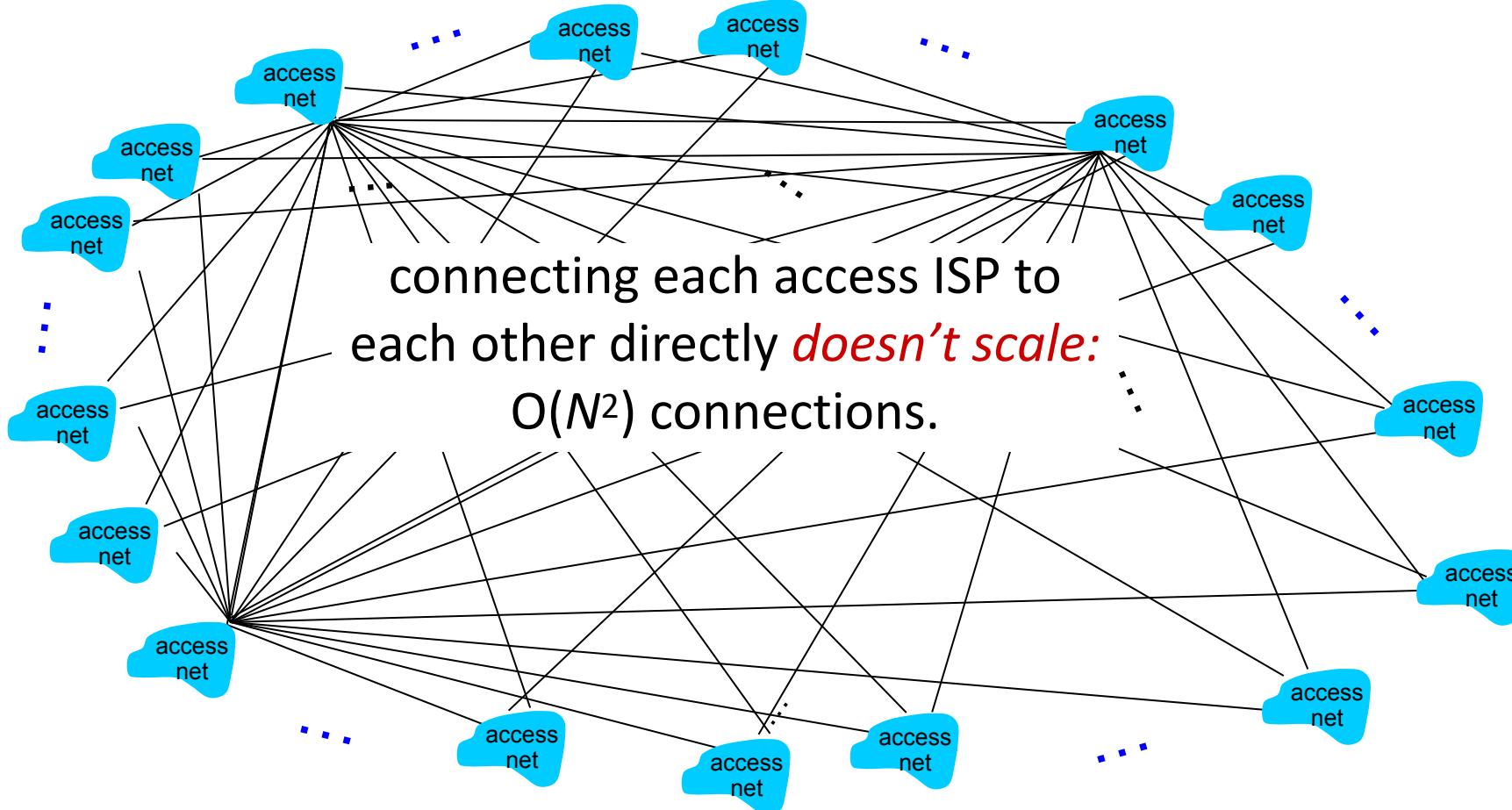
Internet Structure: a “network of networks”

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



Internet Structure: a “network of networks”

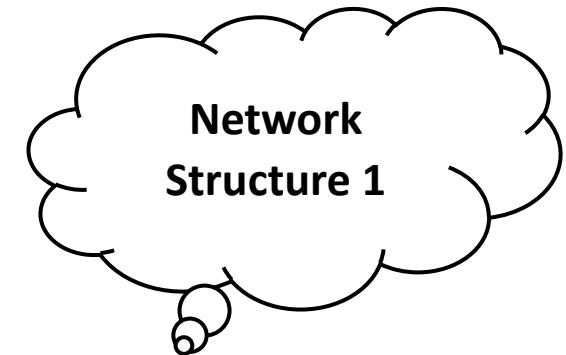
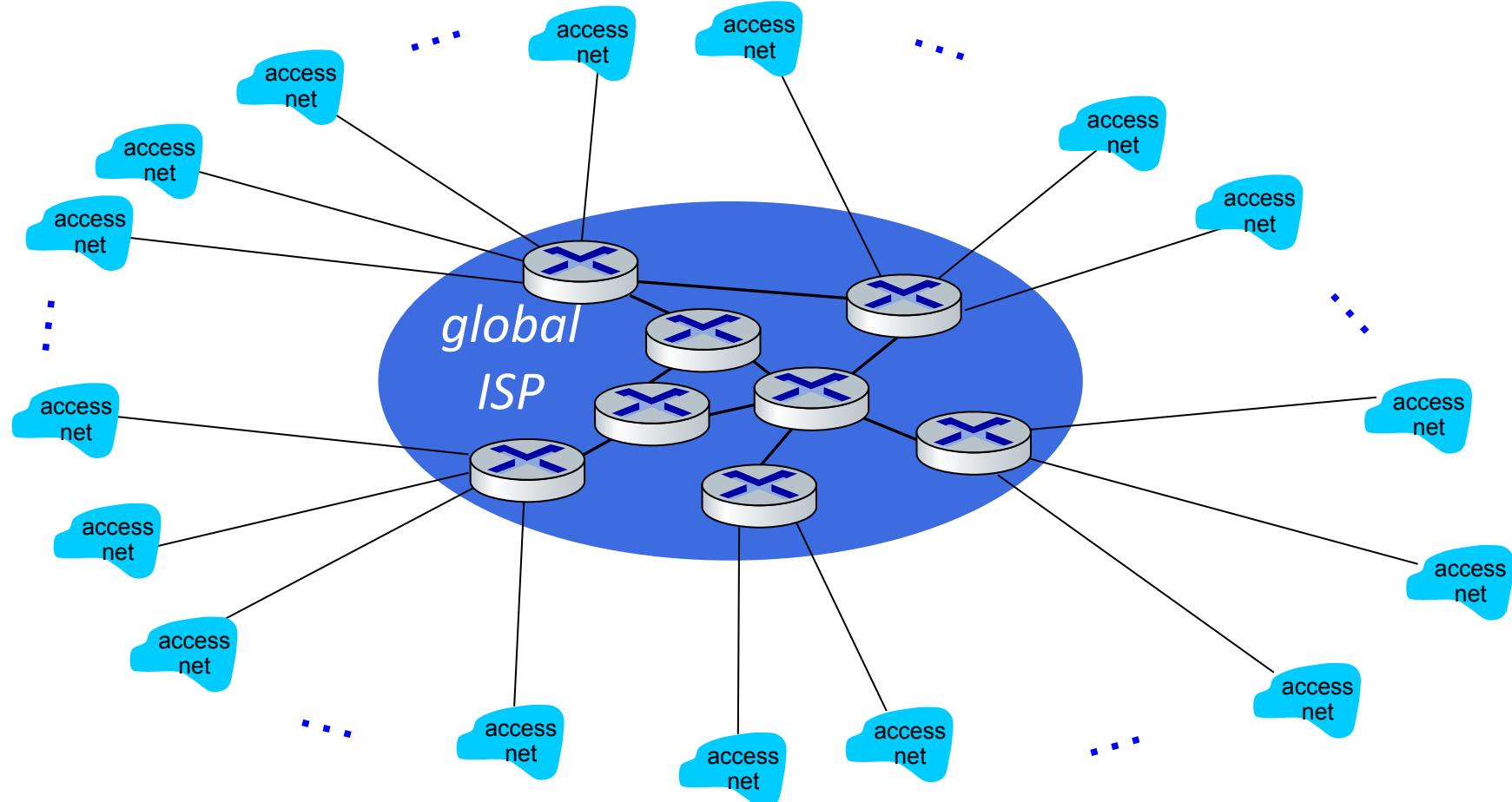
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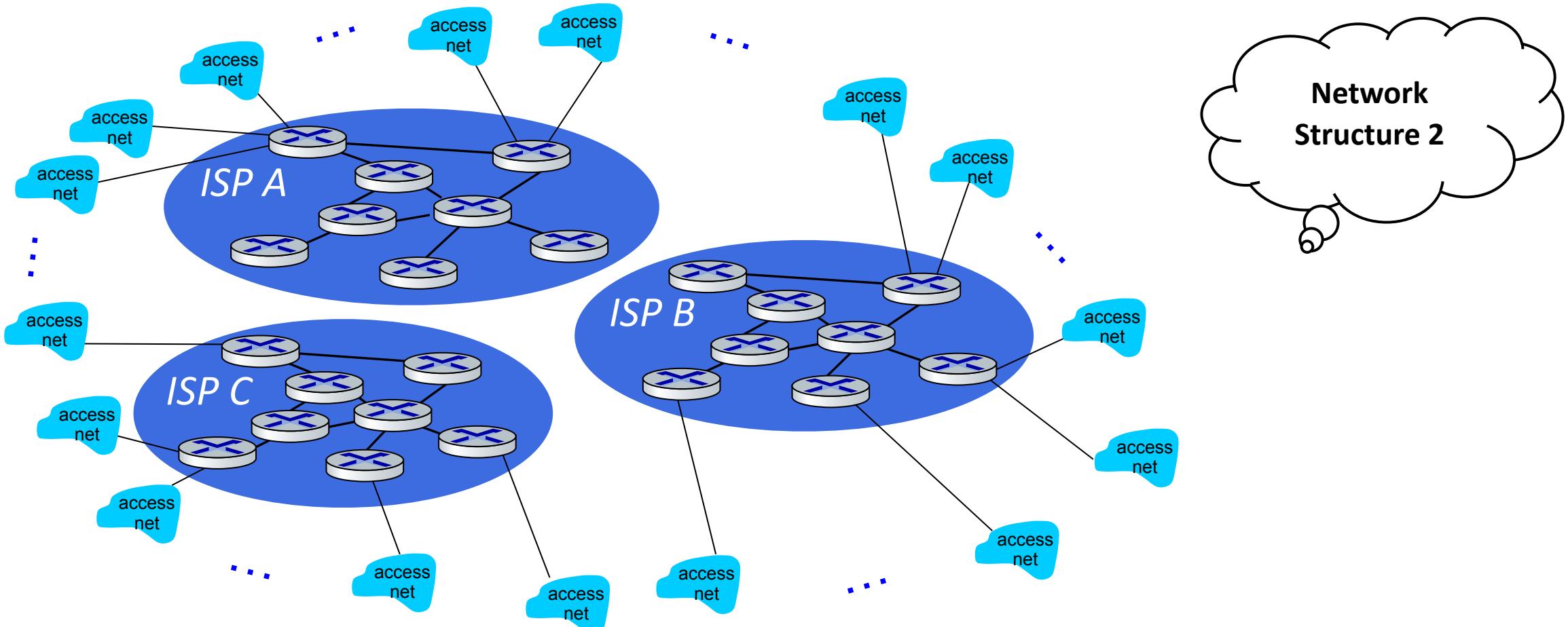
Option: connect each access ISP to one global transit ISP?

Customer and provider ISPs have economic agreement.



Internet Structure: a “network of networks”

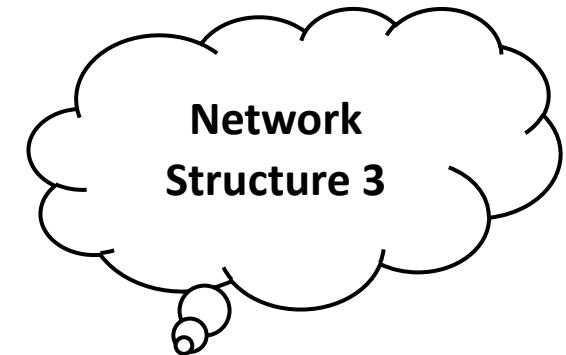
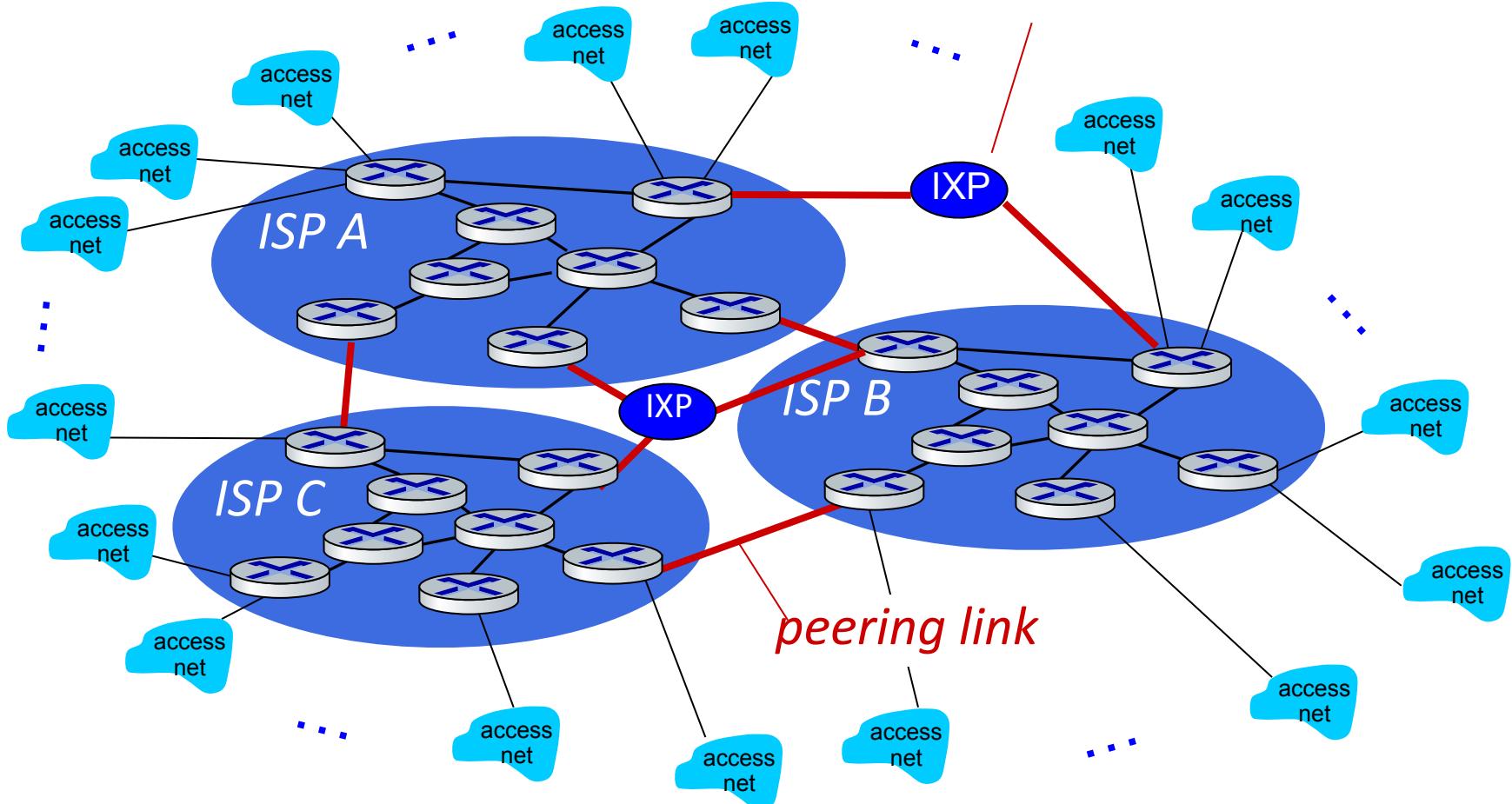
But if one global ISP is viable business, there will be competitors



Internet Structure: a “network of networks”

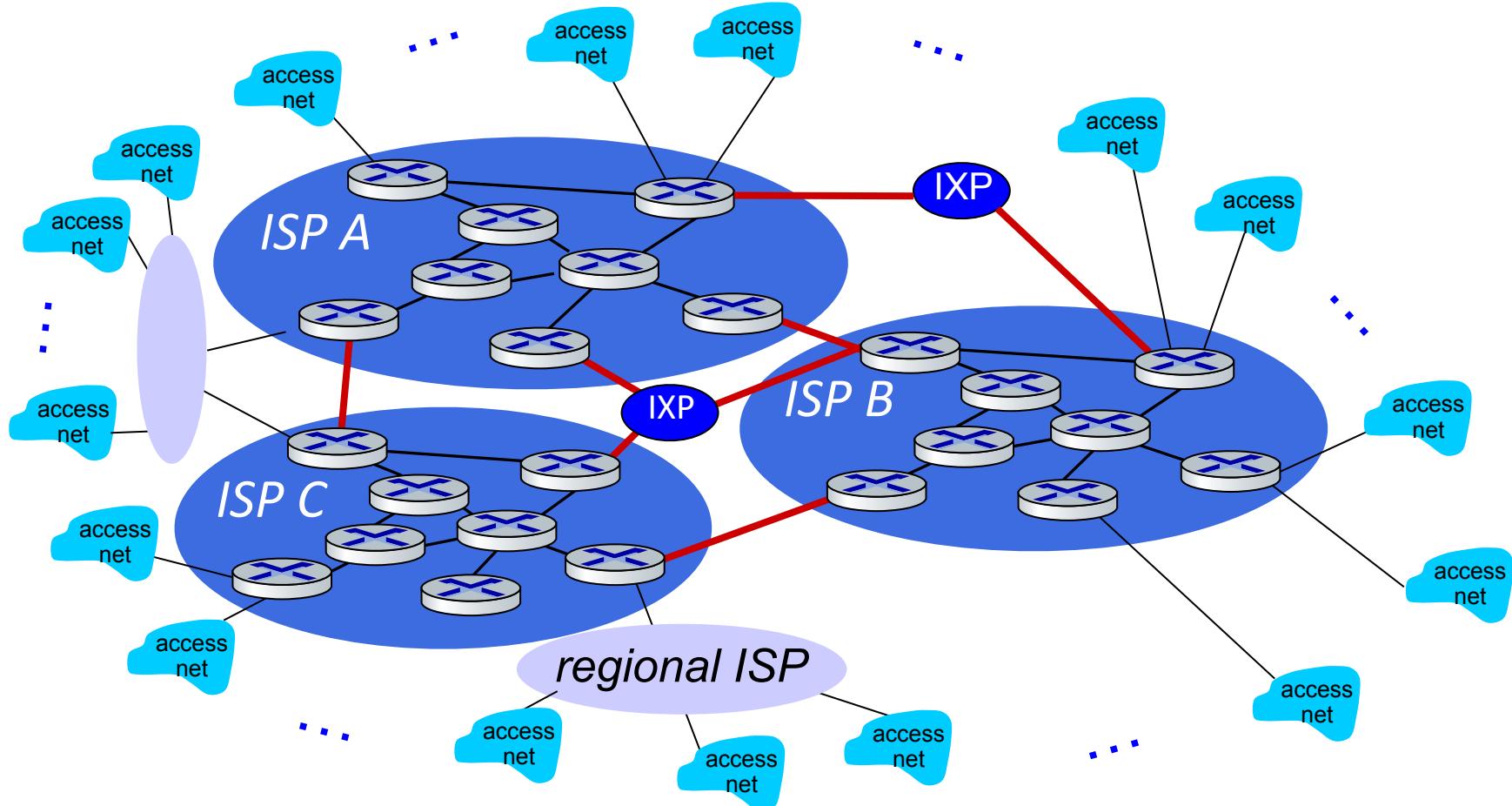
But if one global ISP is viable business, there will be competitors
who will want to be connected

Internet exchange point



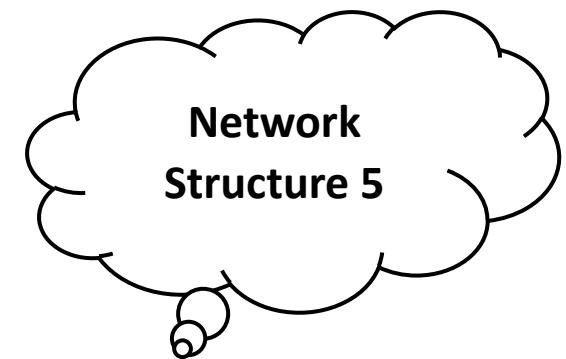
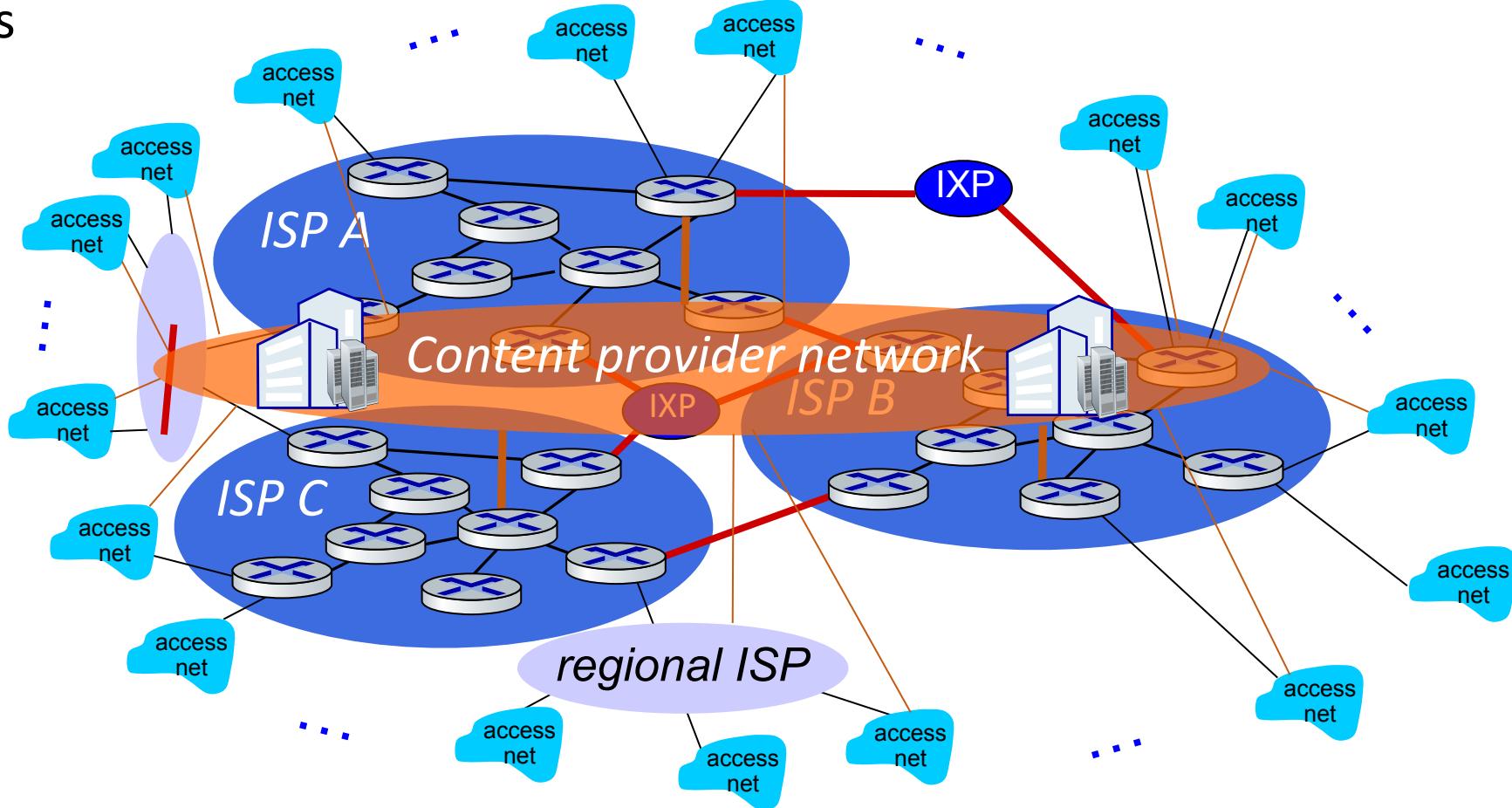
Internet Structure: a “network of networks”

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

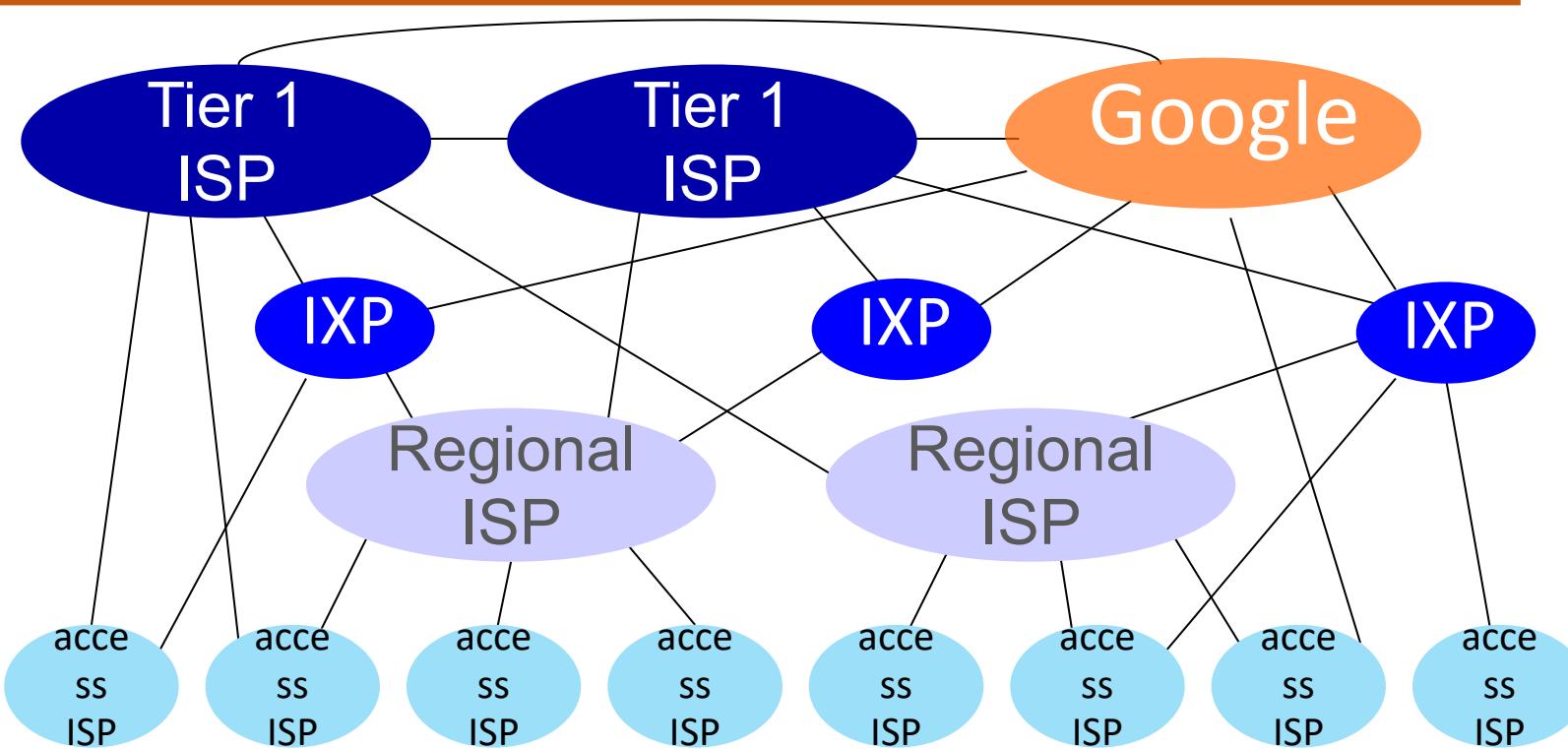


Internet Structure: a “network of networks”

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



Internet Structure: a “network of networks”

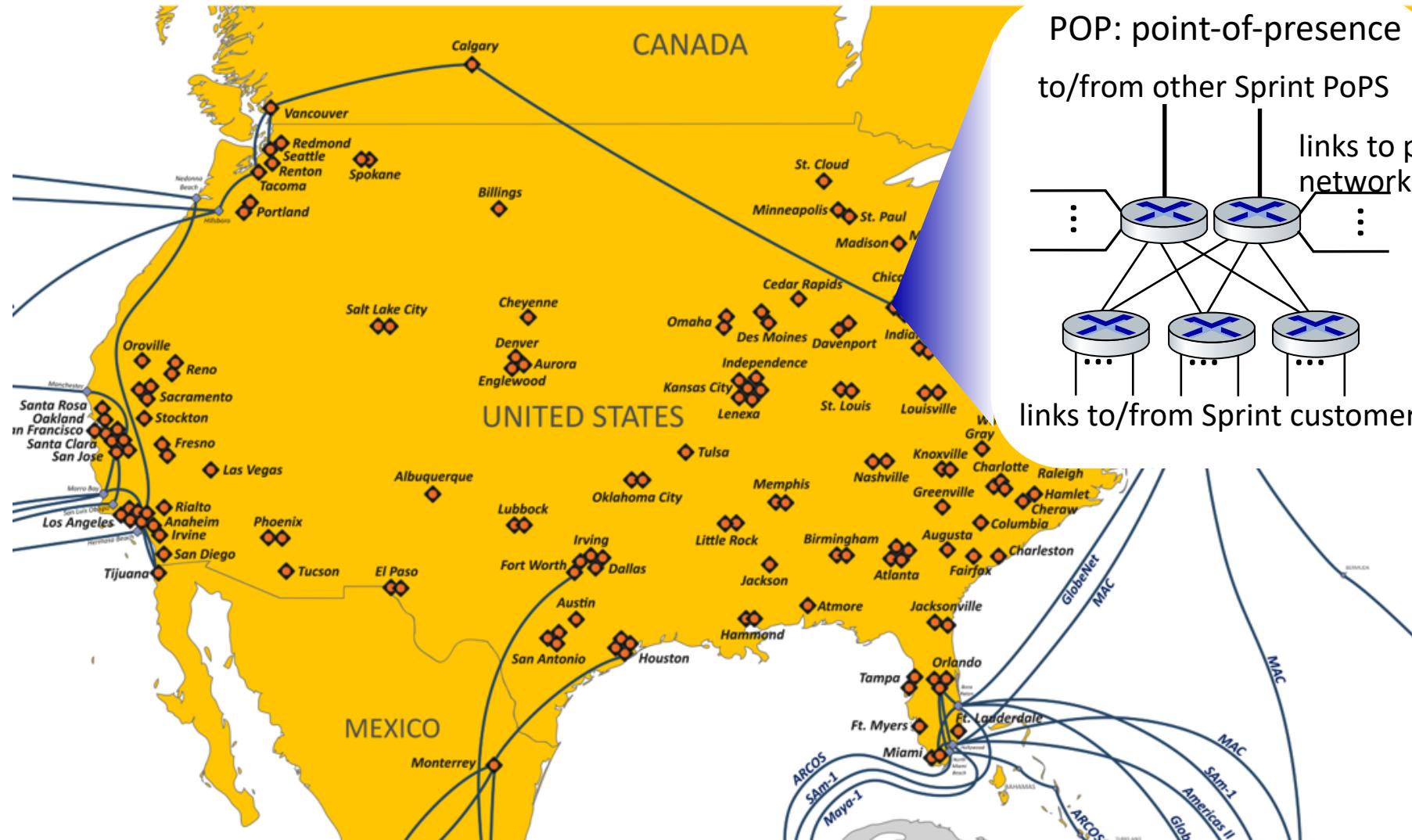


At “center”: small # of well-connected large networks

- “tier-1” commercial ISPs (e.g., Level 3, Sprint, AT&T, NTT), national & international coverage
- content provider networks (e.g., Google, Facebook): private network that connects its data centers to Internet, often bypassing tier-1, regional ISPs

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Network Core: Tier 1 ISP Network Map: Sprint 2019

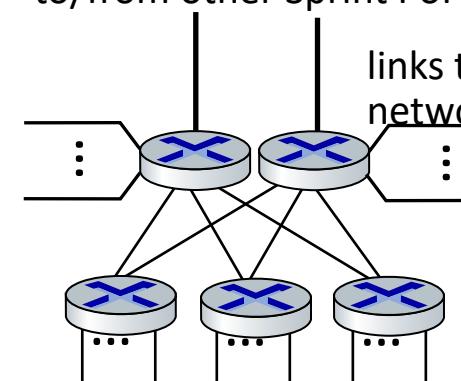


POP: point-of-presence

to/from other Sprint PoPs

links to peering networks

links to/from Sprint customer networks

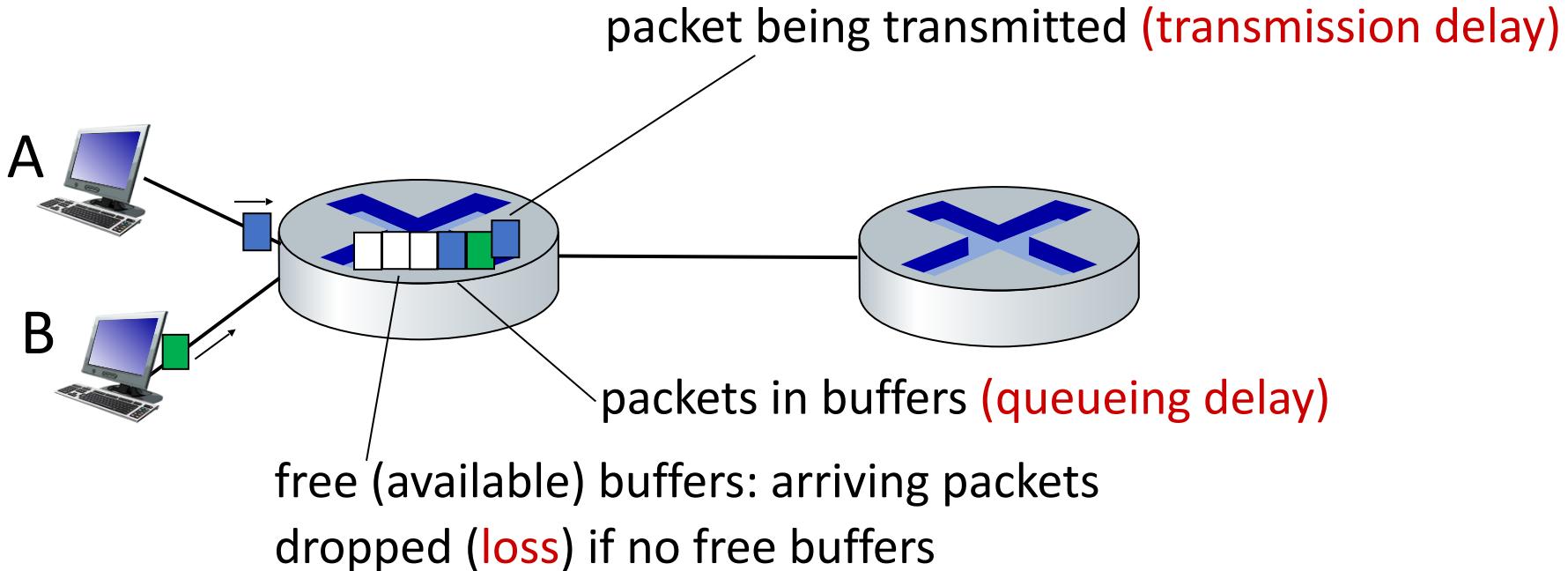


- Sprint Node
 - Sprint Ethernet POP or Sprint Virtual POP
 - Landing Station
 - Sprint Network Backbone
 - Sprint Network Coverage

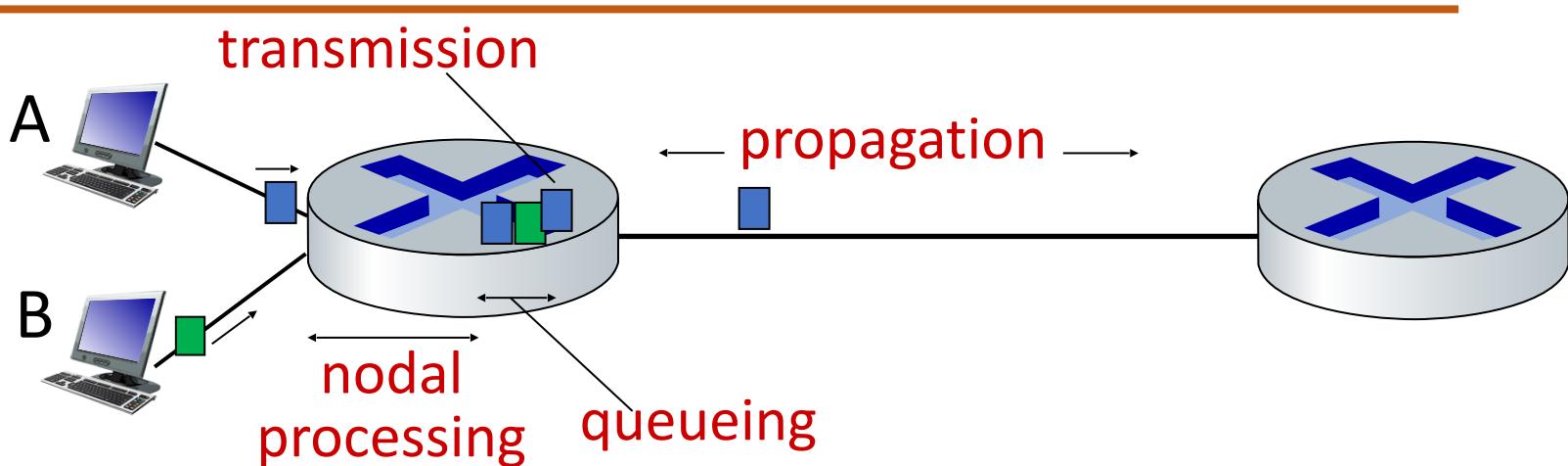
How do packet loss and delay occurs?

packets *queue* in router buffers

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



Performance: Packet Delay – 4 Sources



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

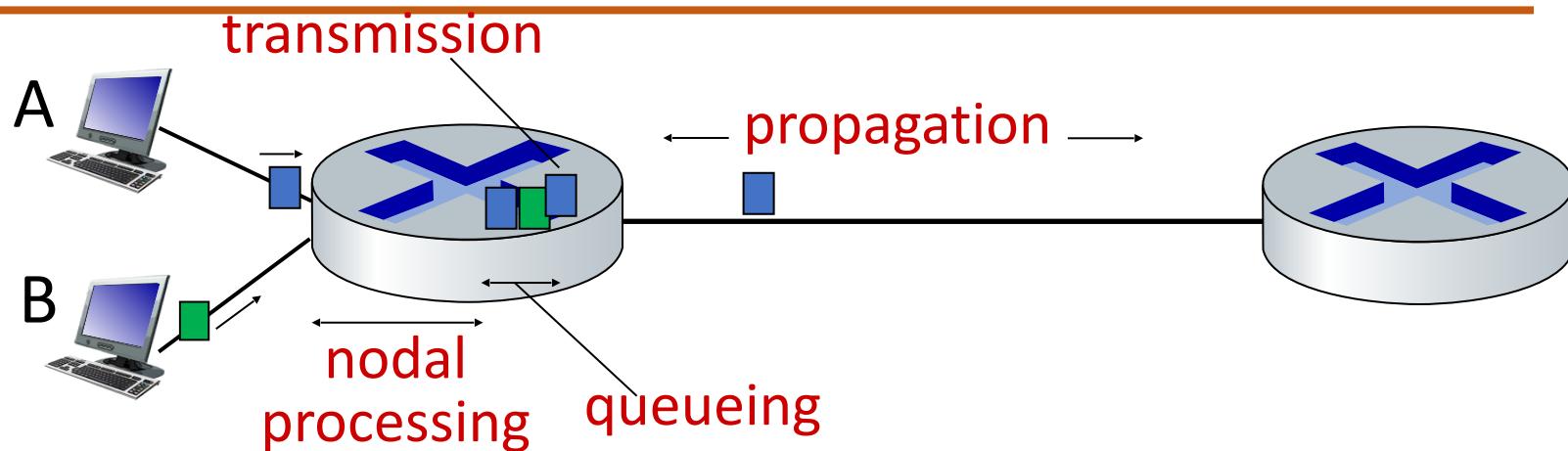
d_{proc} : nodal processing

- check bit errors
- determine output link
- typically < msec

d_{queue} : queueing delay

- time waiting at output link for transmission
- depends on congestion level of router
- microseconds to milliseconds

Performance: Packet Delay – 4 Sources



* Check out the online interactive exercises:
http://gaia.cs.umass.edu/kurose_ross

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

d_{trans} : transmission delay:

- L : packet length (bits)
- R : link transmission rate (bps)
- $d_{\text{trans}} = L/R$

d_{prop} : propagation delay:

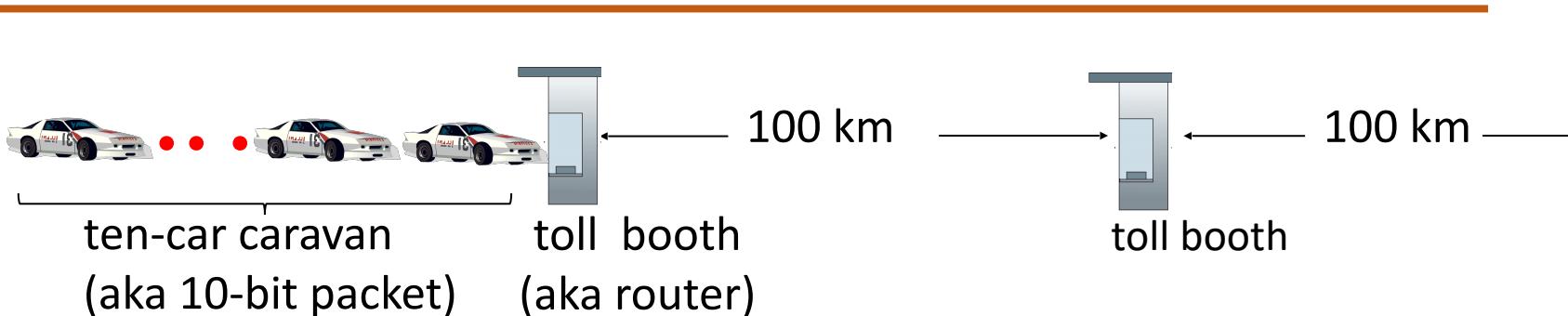
- d : length of physical link
- s : propagation speed ($\sim 2 \times 10^8$ m/sec)
- $d_{\text{prop}} = d/s$

d_{trans} and d_{prop}
very different

Transmission Delay vs Propagation Delay

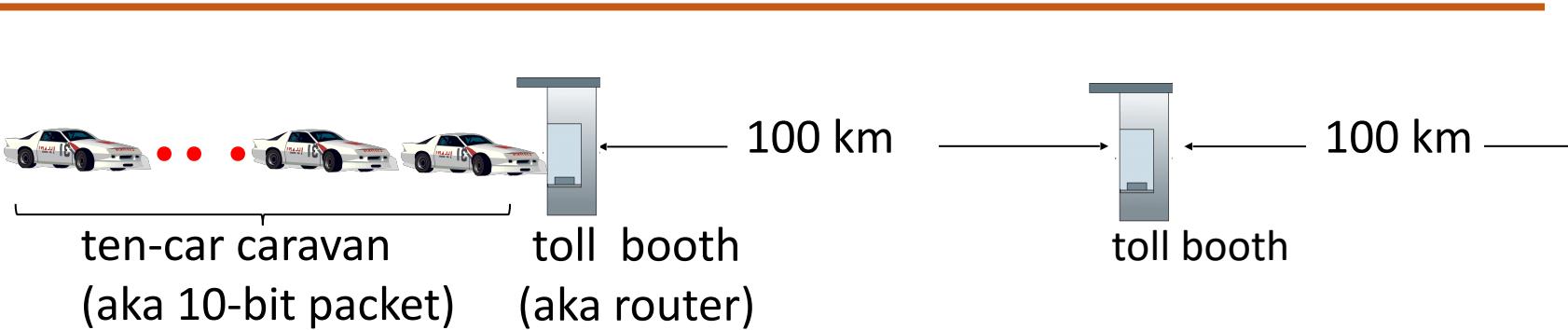
Transmission Delay	Propagation Delay
Time required for the router to push out the packet.	Time it takes a bit to propagate from one router to the next.
A function of the packet's length and the transmission rate of the link.	A function of the distance between the two routers.
$d_{trans} = L/R$	$d_{prop} = d/s$
Nothing to do with the distance between the two routers.	Nothing to do with the packet's length or the transmission rate of the link.

Performance: Delay – Caravan Analogy



- cars “propagate” at 100 km/hr
- toll booth takes 12 sec to service car (bit transmission time)
- car ~ bit; caravan ~ packet
- ***Q: How long until caravan is lined up before 2nd toll booth?***
- time to “push” entire caravan through toll booth onto highway = $12*10 = 120$ sec
- time for last car to propagate from 1st to 2nd toll both:
 $100\text{km}/(100\text{km/hr}) = 1 \text{ hr}$
- ***A: 62 minutes***

Performance: Delay – Caravan Analogy (more)



- suppose cars now “propagate” at 1000 km/hr
- and suppose toll booth now takes one min to service a car
- *Q: Will cars arrive to 2nd booth before all cars serviced at first booth?*

A: Yes! after 7 min, first car arrives at second booth; three cars still at first booth

Unlike other delays (dproc, dtrans, dprop), dqueue is interesting.

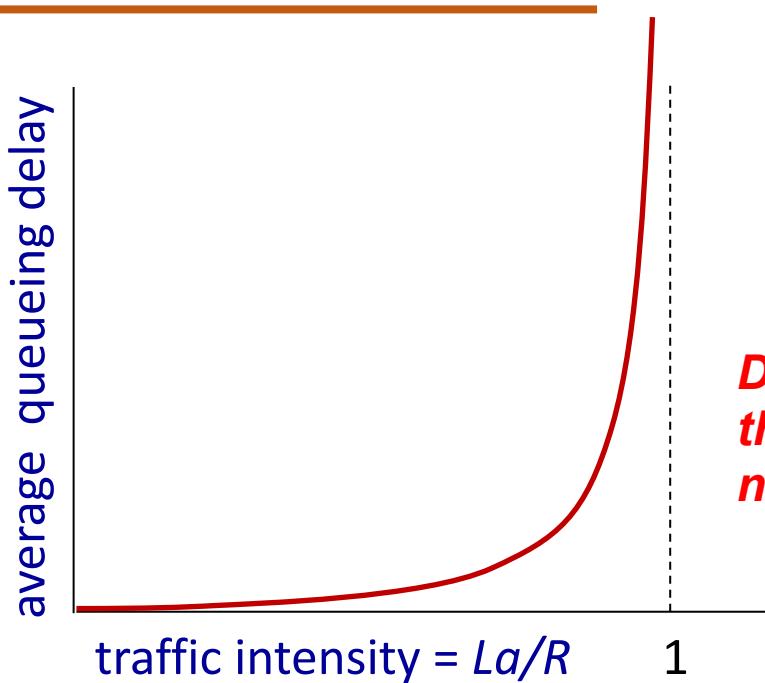
- Can vary from packet to packet.
- Characterize d_{queue} -> average, variance, probability that it exceeds some specified value.

When is the queuing delay large and when is it insignificant?

- Rate at which traffic arrives at the queue,
- Transmission rate of the link,
- Nature of the arriving traffic – periodically or in bursts

Performance: Packet Queueing Delay revisited

- R : link bandwidth (bps)
- L : packet length (bits)
- a : average packet arrival rate (pps)
- La : avg. rate at which bits arrive at the queue
- $La/R > 1$: more “work” arriving is more than can be serviced - average delay infinite!
- $La/R \leq 1$: nature of arriving traffic
- $La/R \sim 0$: avg. queueing delay small



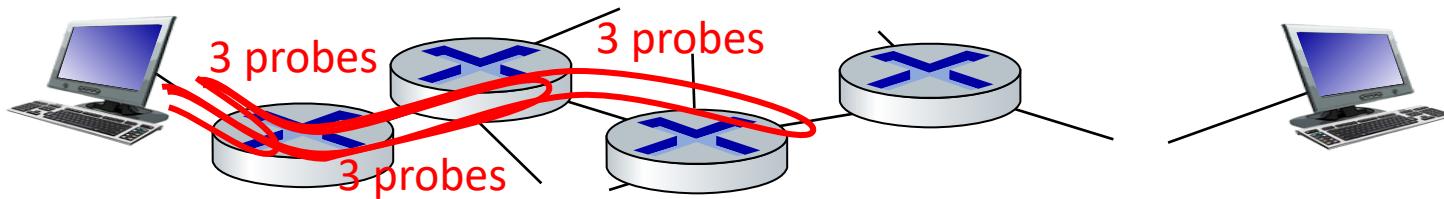
Design your system so that the traffic intensity is no greater than 1.



$La/R > 1$: Average rate at which bits arrive at the queue exceeds the rate at which the bits can be transmitted from the queue.

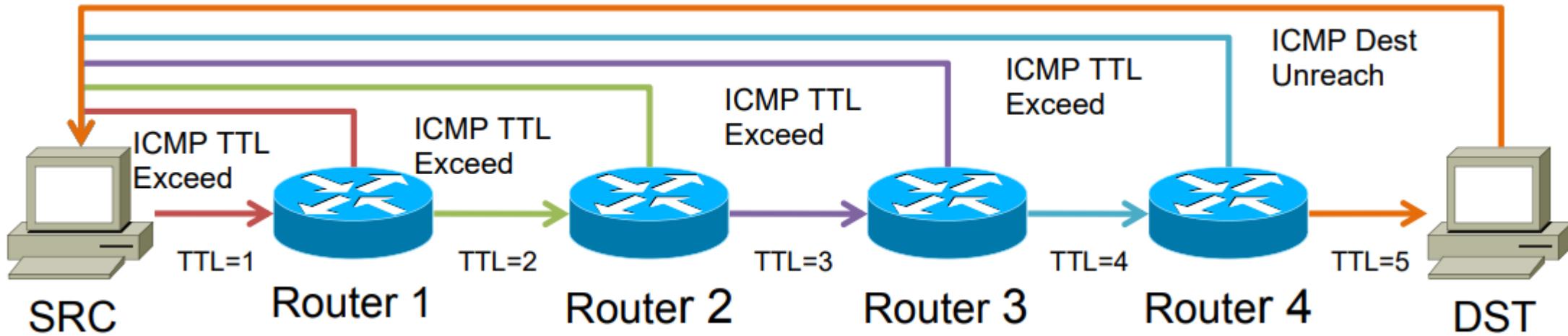
“Real” Internet Delays and Routes

- 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 (with time-to-live field value of i)
 - router i will return packets to sender
 - sender measures time interval between transmission and reply



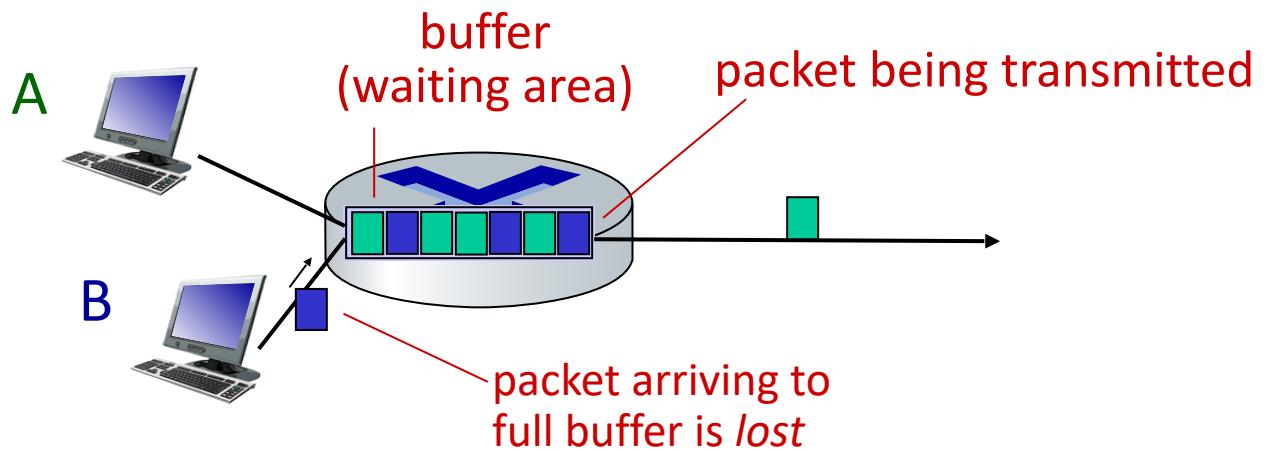
traceroute: gaia.cs.umass.edu to www.eurecom.fr

		3 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	***	* means no response (probe lost, router not replying)
19	fantasia.eurecom.fr (193.55.113.142)	132 ms 128 ms 136 ms



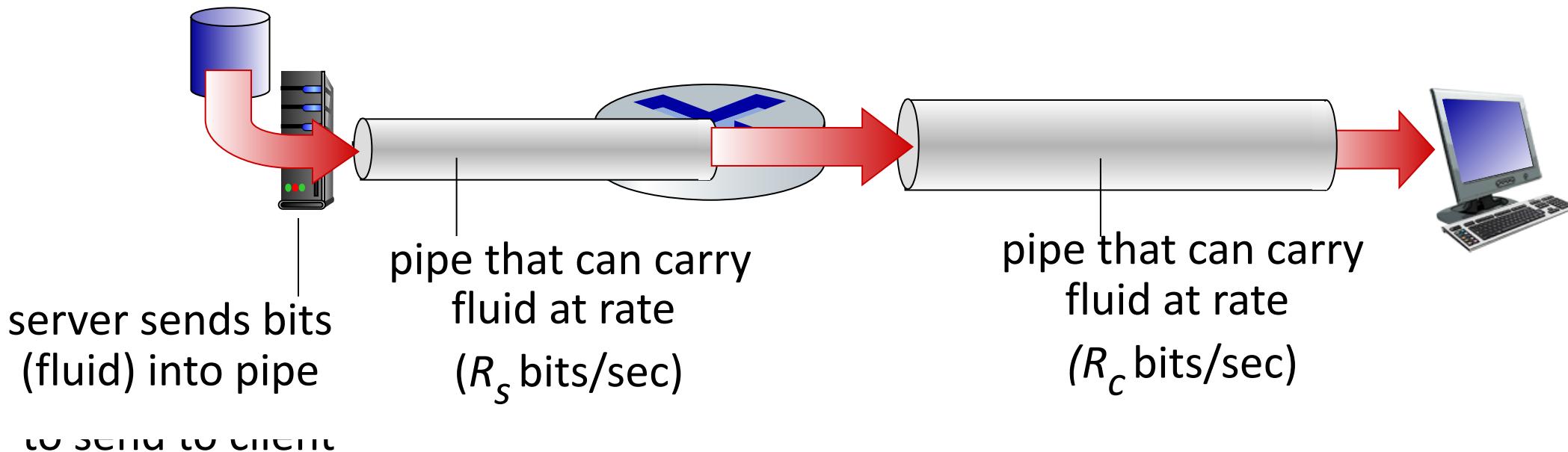
Refer RFC 1393, **Traceroute Using an IP Option**
Don't Trust Traceroute (Completely)

- queue (aka buffer) preceding link in buffer has finite capacity
- packet arriving to full queue dropped (aka lost)
- lost packet may be retransmitted by previous node, by source end system, or not at all



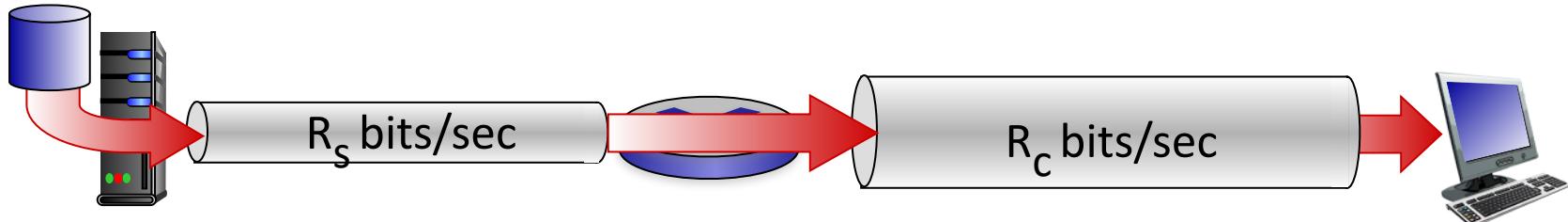
* Check out the Java applet for an interactive animation on queuing and loss

- **throughput:** rate (bits/time unit) at which bits are being sent from sender to receiver
 - *instantaneous:* rate at given point in time
 - *average:* rate over longer period of time



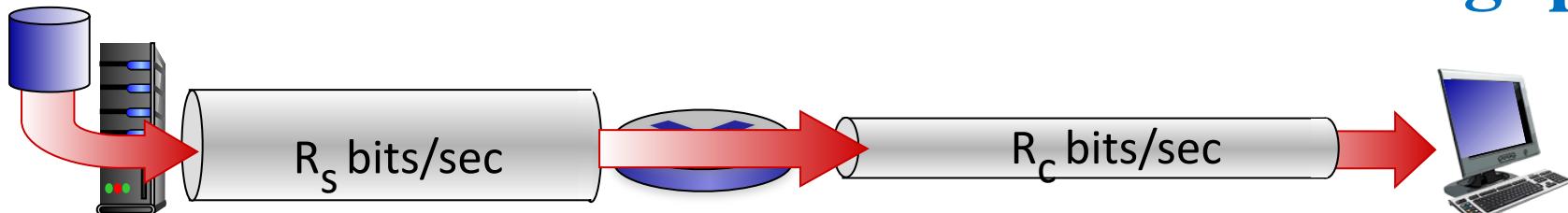
Performance: Throughput (more)

$R_s < R_c$ What is average end-end throughput?



$R_s > R_c$ What is average end-end throughput?

$$\text{Throughput} = \min\{R_s, R_c\}$$



bottleneck link

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

Throughput – Numerical Example

- Suppose you are downloading an MP3 file of $F = 32$ million bits.
- The server has a transmission rate of $R_s = 2$ Mbps and you have an access link of $R_c = 1$ Mbps.
- What is the time needed to transfer the file?

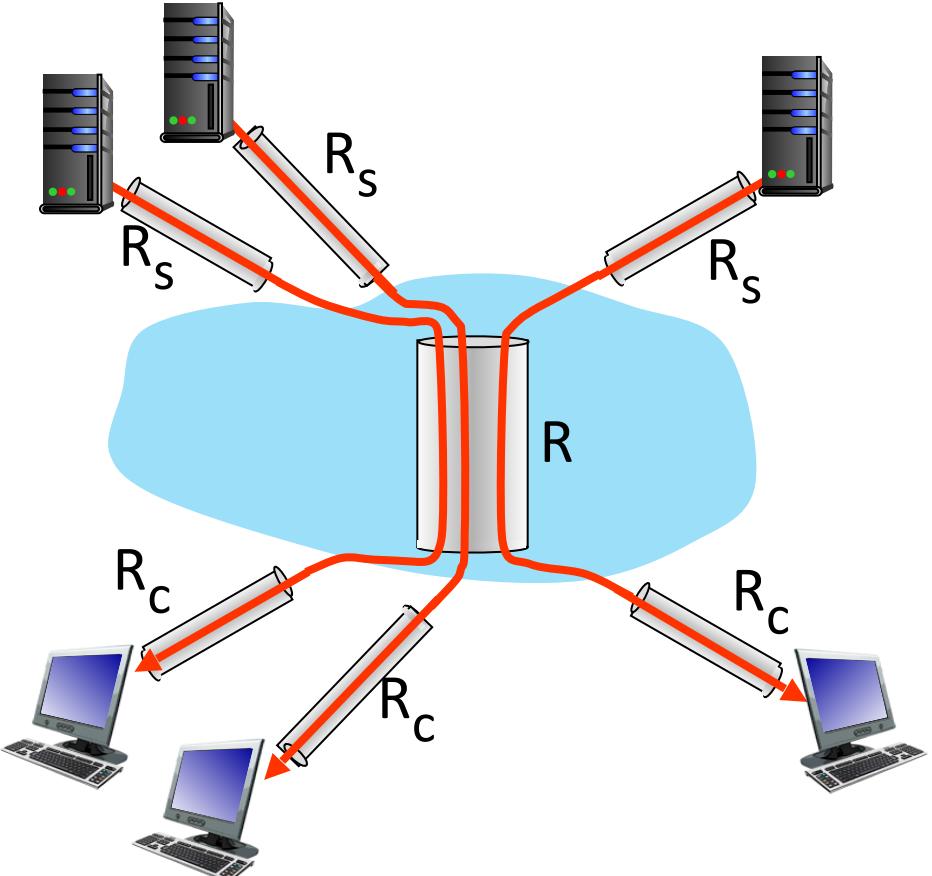
Let's work it out!

Solution:

- 32 seconds!



Performance: Throughput – Network Scenario



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

- per-connection end-end throughput: $\min(R_c, R_s, R/10)$
- in practice: R_c or R_s is often bottleneck

* Check out the online interactive exercises for more examples: http://gaia.cs.umass.edu/kurose_ross/

- Suppose $R_s = 2$ Mbps, $R_c = 1$ Mbps, $R = 5$ Mbps
- 10 clients from 10 servers = 10 downloads

End-to-end throughput for each download is now reduced to 500 kbps.



Computer Networks and the Internet

Protocol Layers

Team Networks

Department of Computer Science and Engineering

Networks are complex,
with many “pieces”:

- hosts
- routers
- links of various media
- applications
- protocols
- hardware, software

Question:

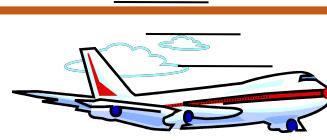
is there any hope of
organizing structure of
network?

.... or at least our *discussion*
of networks?

COMPUTER NETWORKS

Example: Organization of Air Travel

ticket (purchase)
baggage (check)
gates (load)
runway takeoff
airplane routing



ticket (complain)
baggage (claim)
gates (unload)
runway landing
airplane routing

airplane routing

airline travel: a series of steps, involving many services

Layering of Airline functionality



layers: each layer implements a service

- via its own internal-layer actions
- relying on services provided by layer below

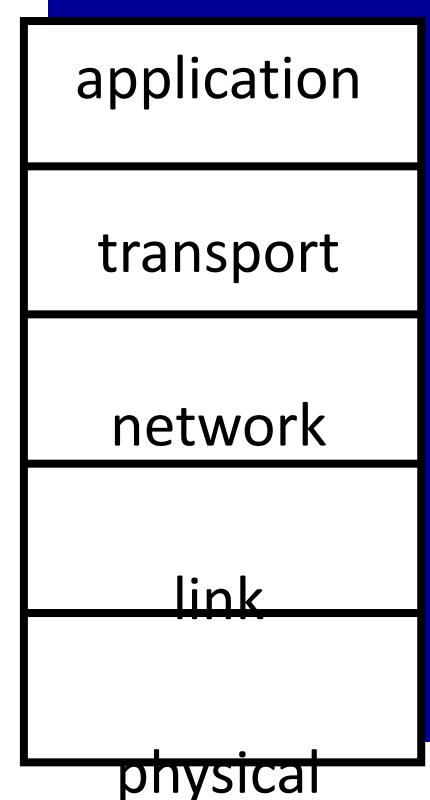
Q: describe in words the service provided in each layer above

Why layering?

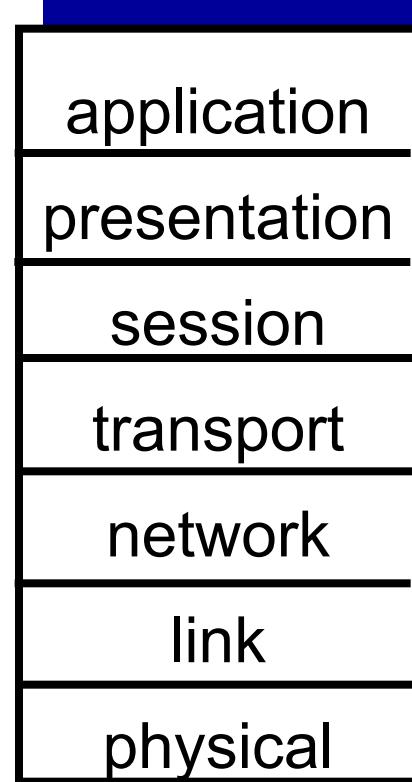
dealing with complex systems:

- explicit structure allows identification, relationship of complex system's pieces
 - layered *reference model* for discussion
- modularization eases maintenance, updating of system
 - change in layer's service *implementation*: transparent to rest of system
 - e.g., change in gate procedure doesn't affect rest of system
- layering considered harmful?
- layering in other complex systems?

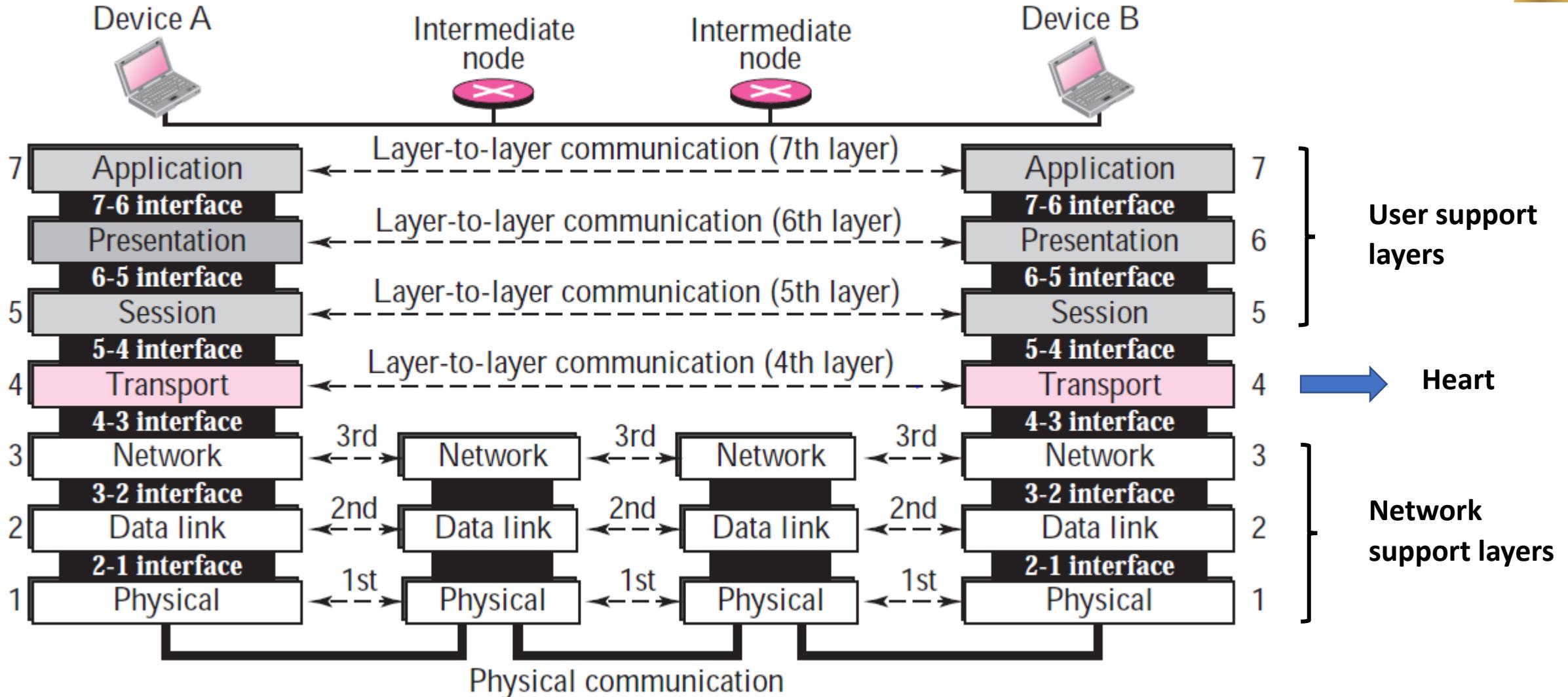
- ***application***: supporting network applications (access to network resources)
 - IMAP, SMTP, HTTP
- ***transport***: process-process data transfer (segmentation & reassembly, sockets, connection, flow and error control)
 - TCP, UDP
- ***network***: routing of datagrams from source to destination (addressing, routing)
 - IP, routing protocols
- ***link***: data transfer between neighboring network elements (framing, addressing, flow & error control)
 - Ethernet, 802.11 (WiFi), PPP
- ***physical***: bits “on the wire”



- ***presentation:*** allow applications to interpret meaning of data, (e.g., encryption, compression, machine-specific conventions)
- ***session:*** synchronization, checkpointing, recovery of data exchange
- Internet stack “missing” these layers!
 - these services, *if needed*, must be implemented in application
 - needed?

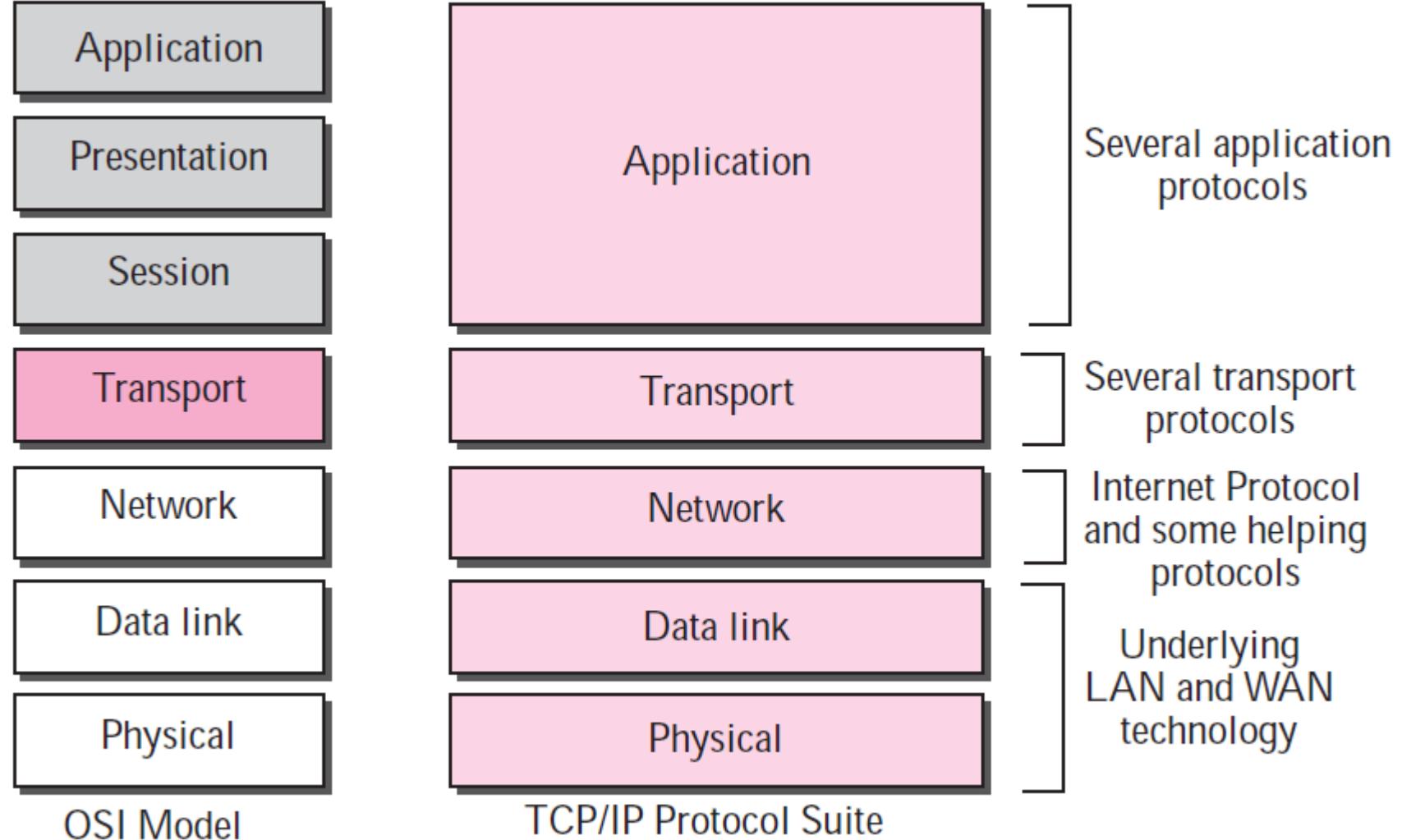


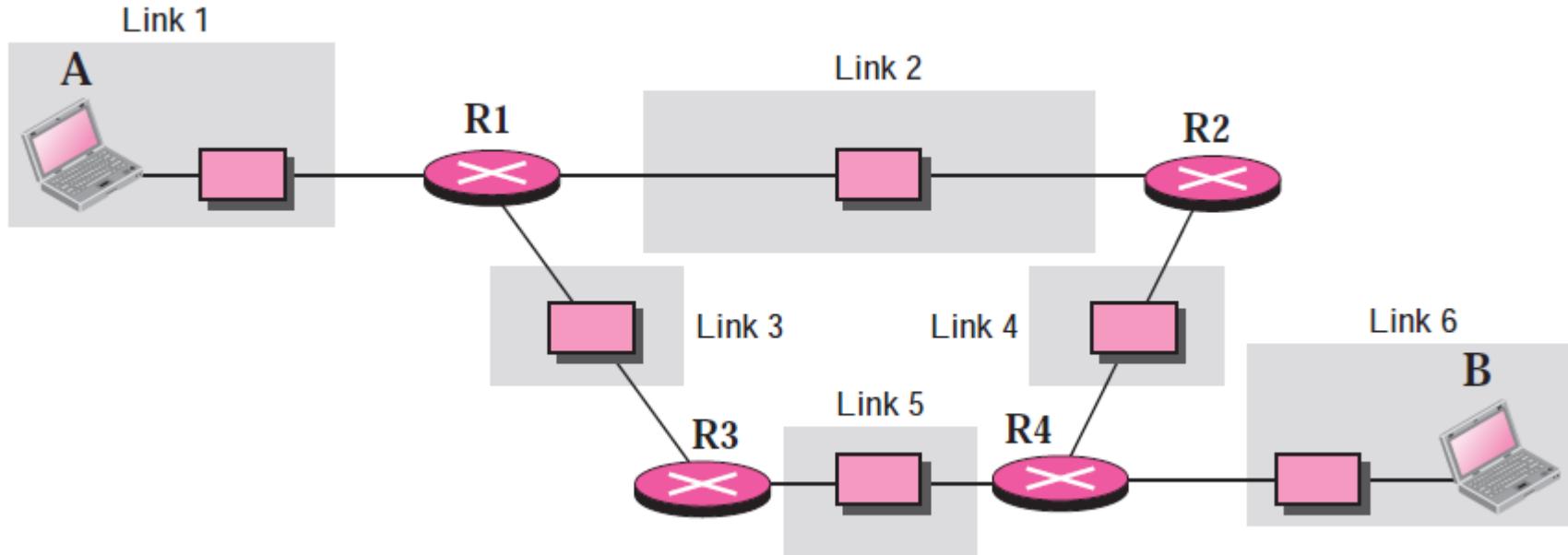
Open Systems Interconnection (OSI) model – introduced in late 1970s by ISO.



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TCP/IP vs OSI reference model

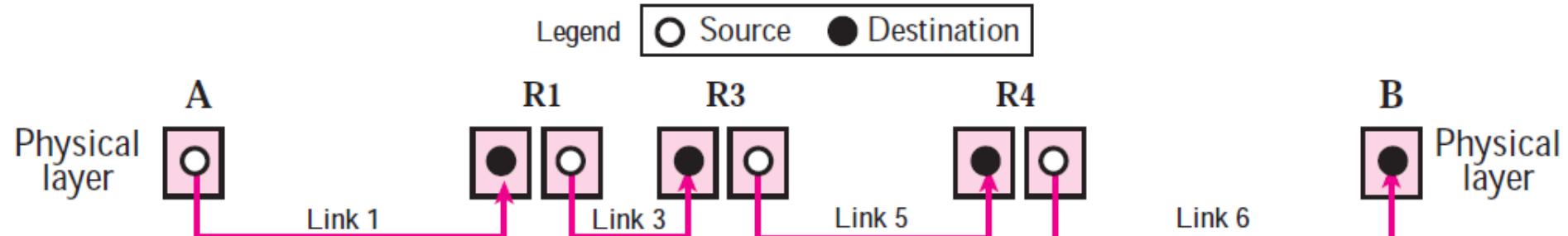




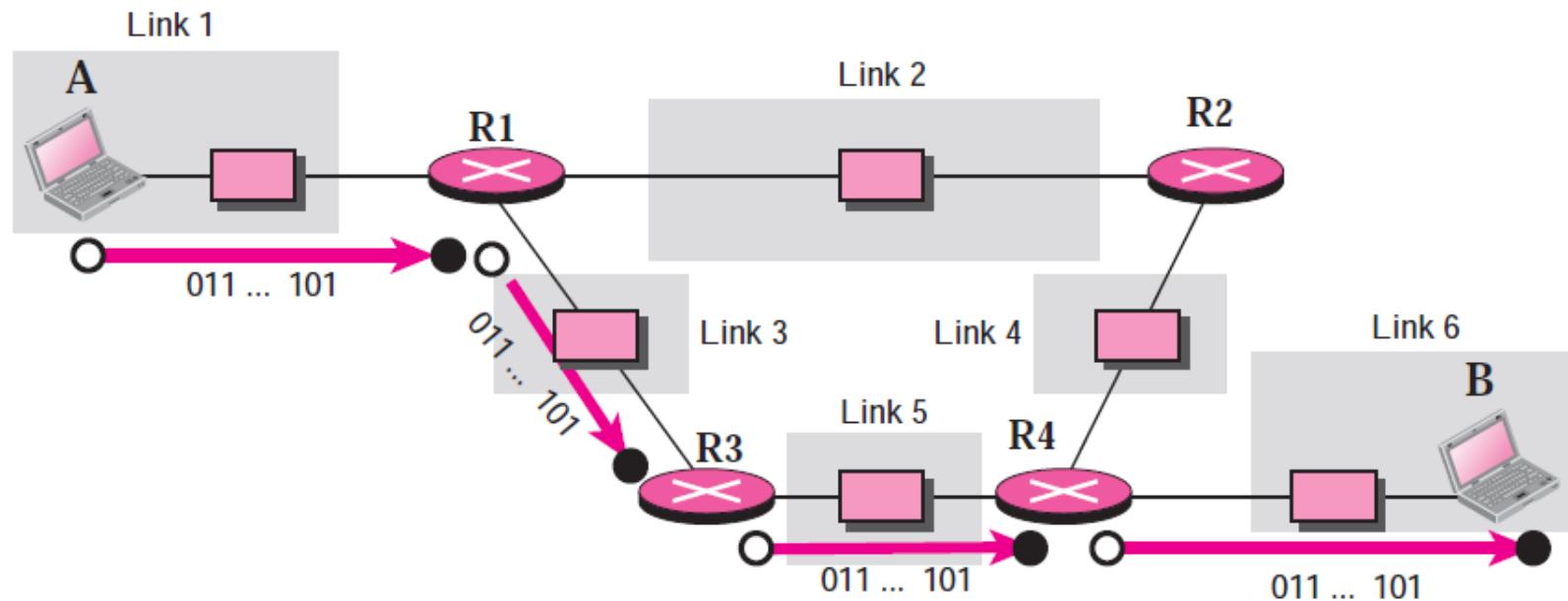
A private internet

COMPUTER NETWORKS

Layers in the TCP/IP Protocol Suite (more)



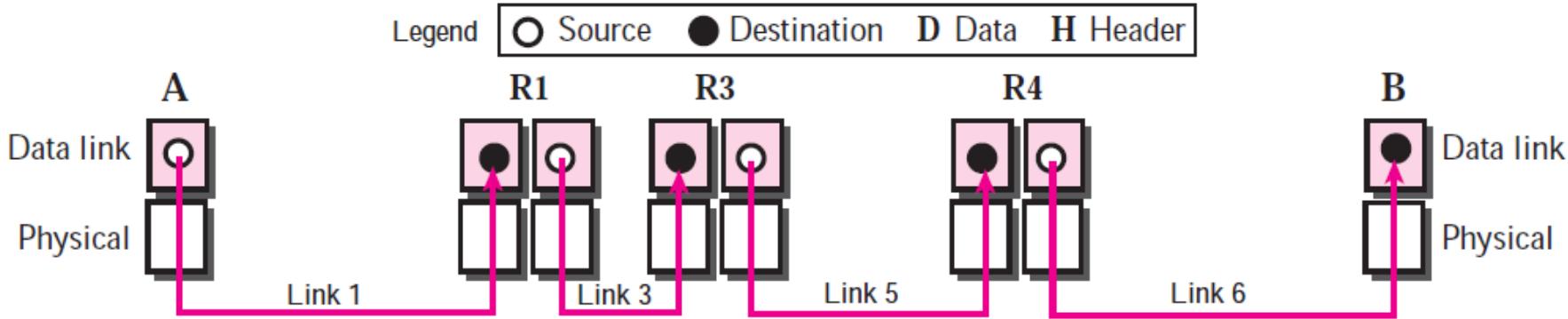
Communication at the physical layer



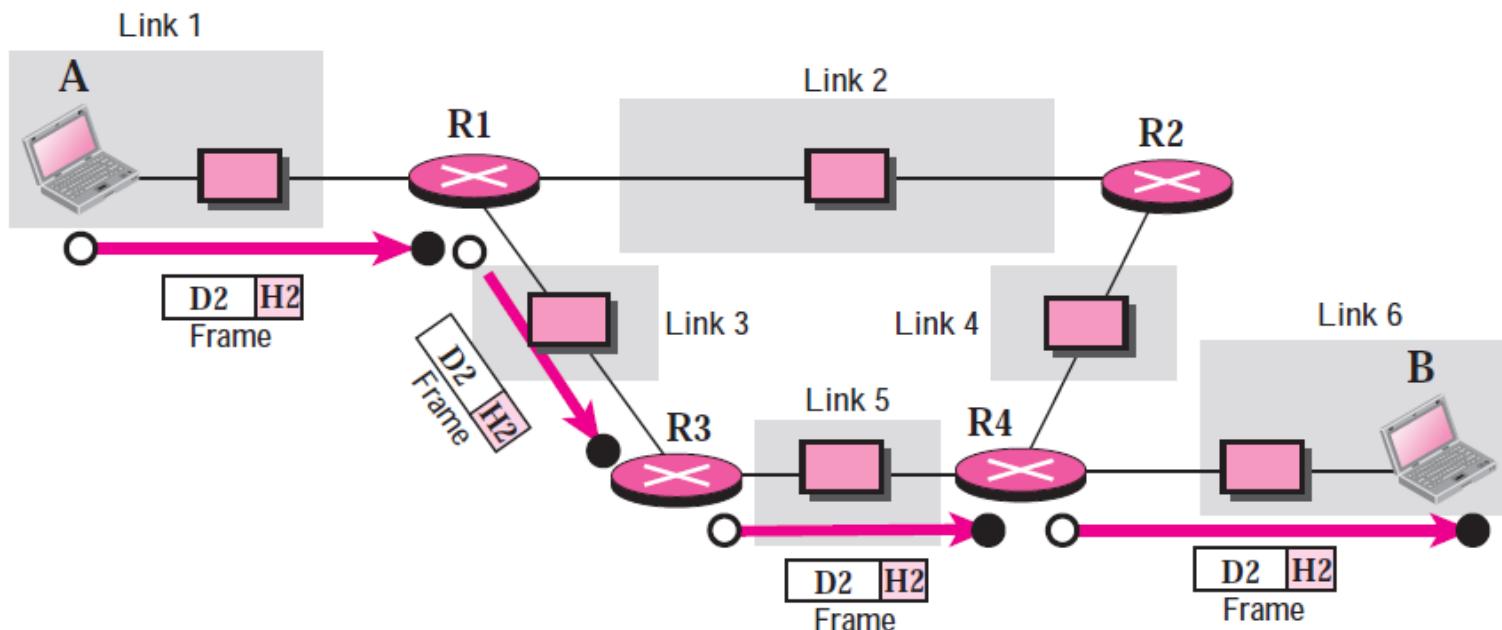
Unit of Communication

- bit

Layers in the TCP/IP Protocol Suite (more)

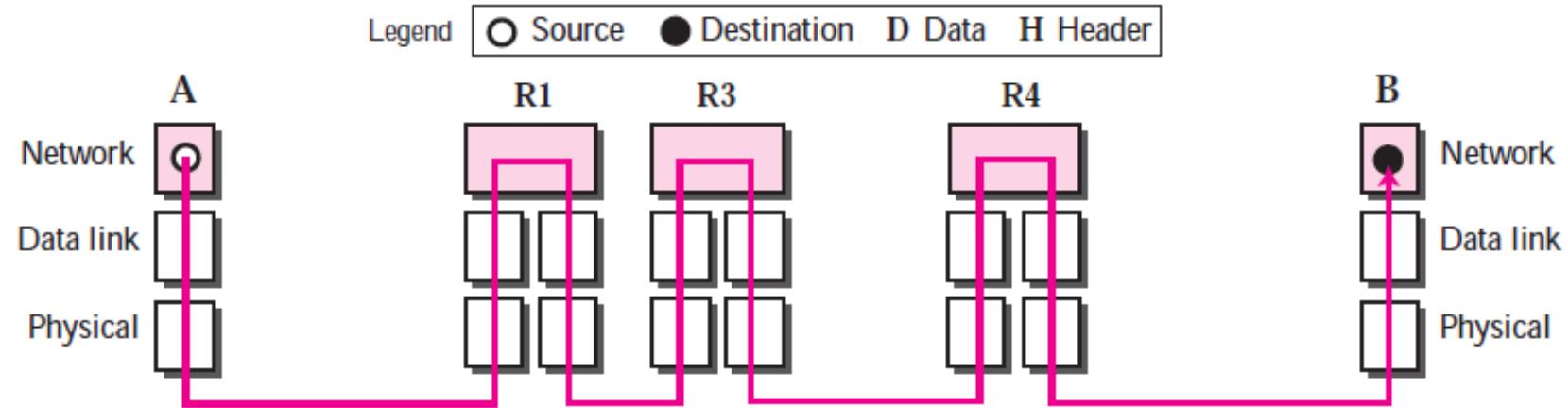


Communication at the data link layer

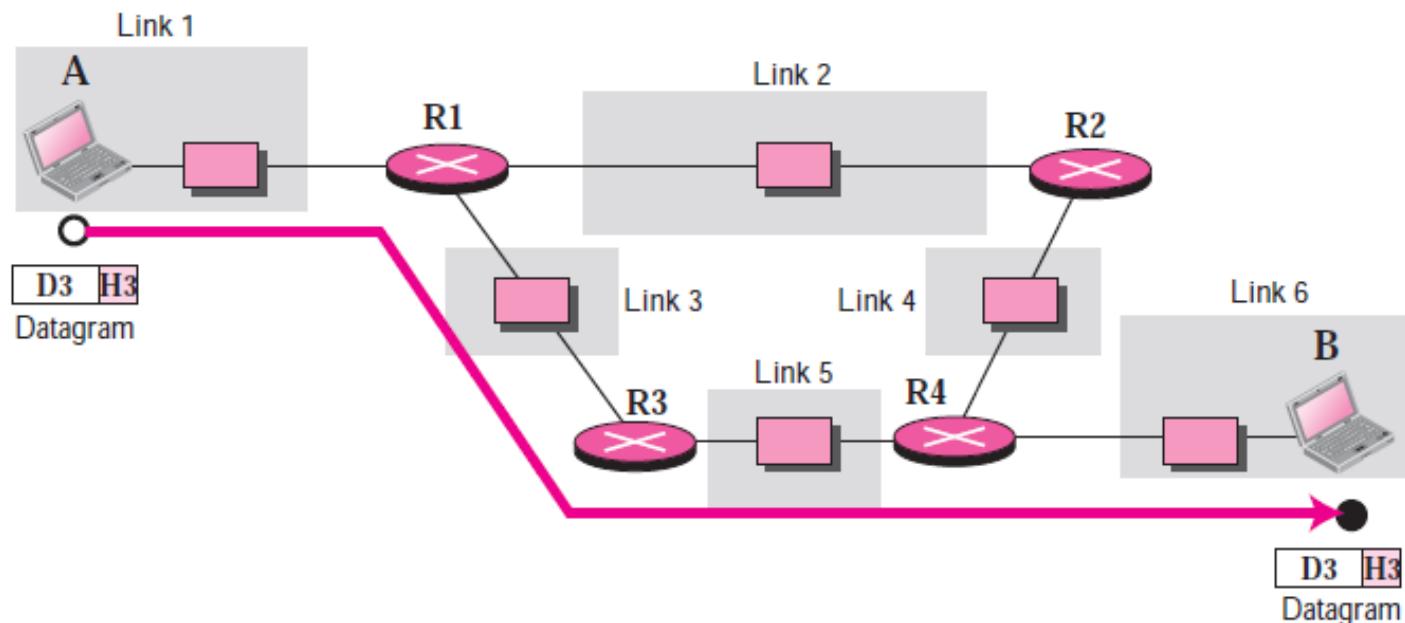


Unit of Communication – frame

Layers in the TCP/IP Protocol Suite (more)



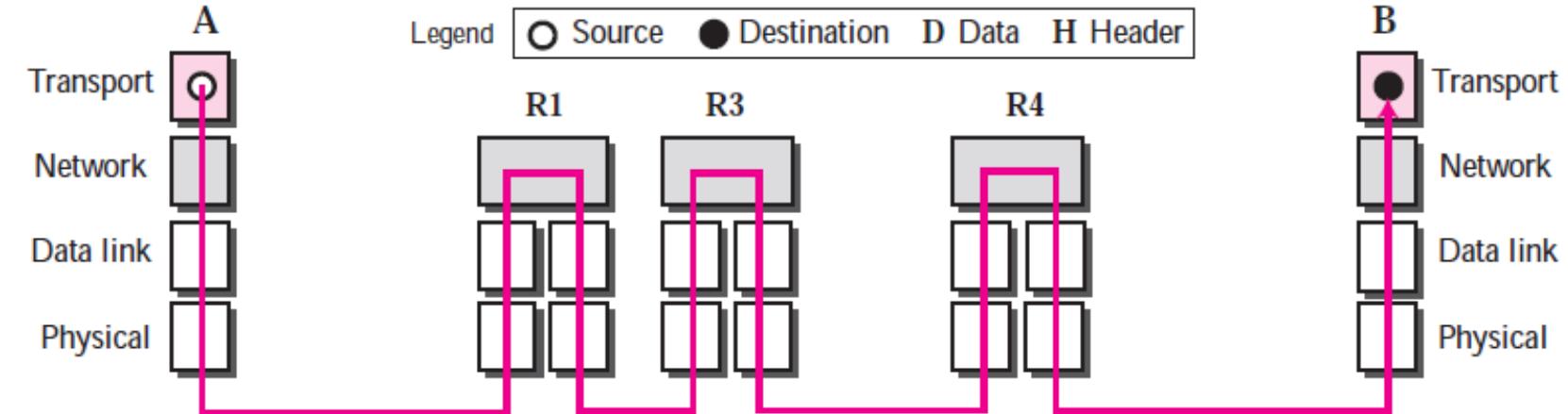
Communication at the network layer



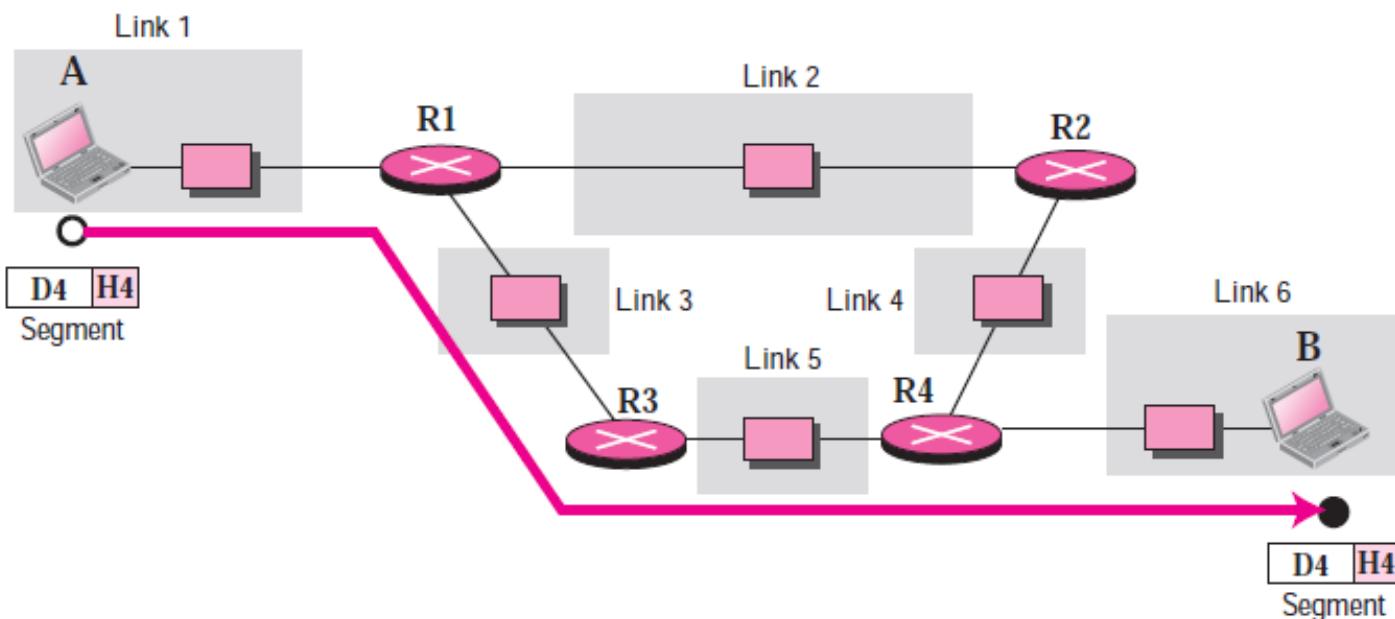
Unit of Communication –
datagram

COMPUTER NETWORKS

Layers in the TCP/IP Protocol Suite (more)

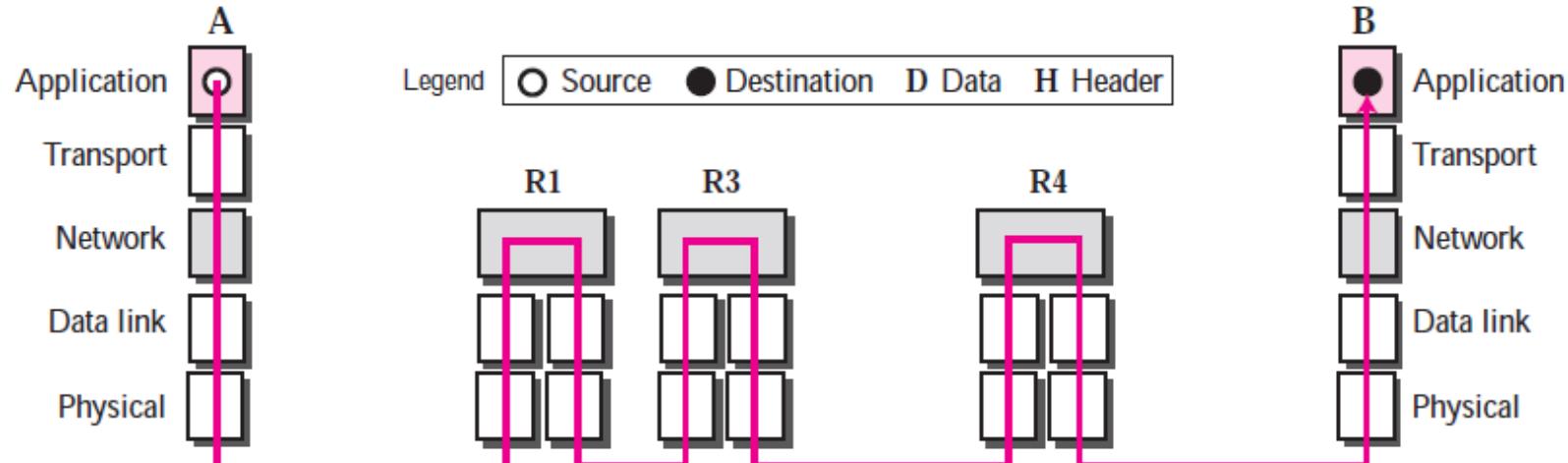


Communication at the transport layer

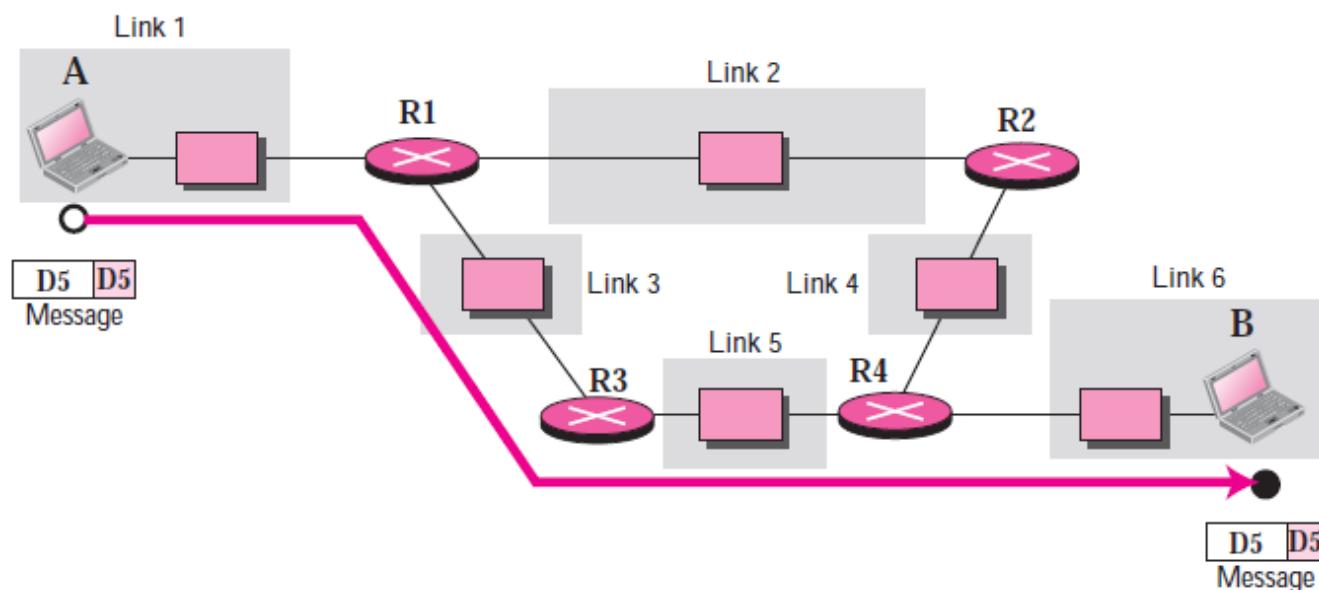


Unit of Communication
– segment/packet

Layers in the TCP/IP Protocol Suite (more)



Communication at the application layer

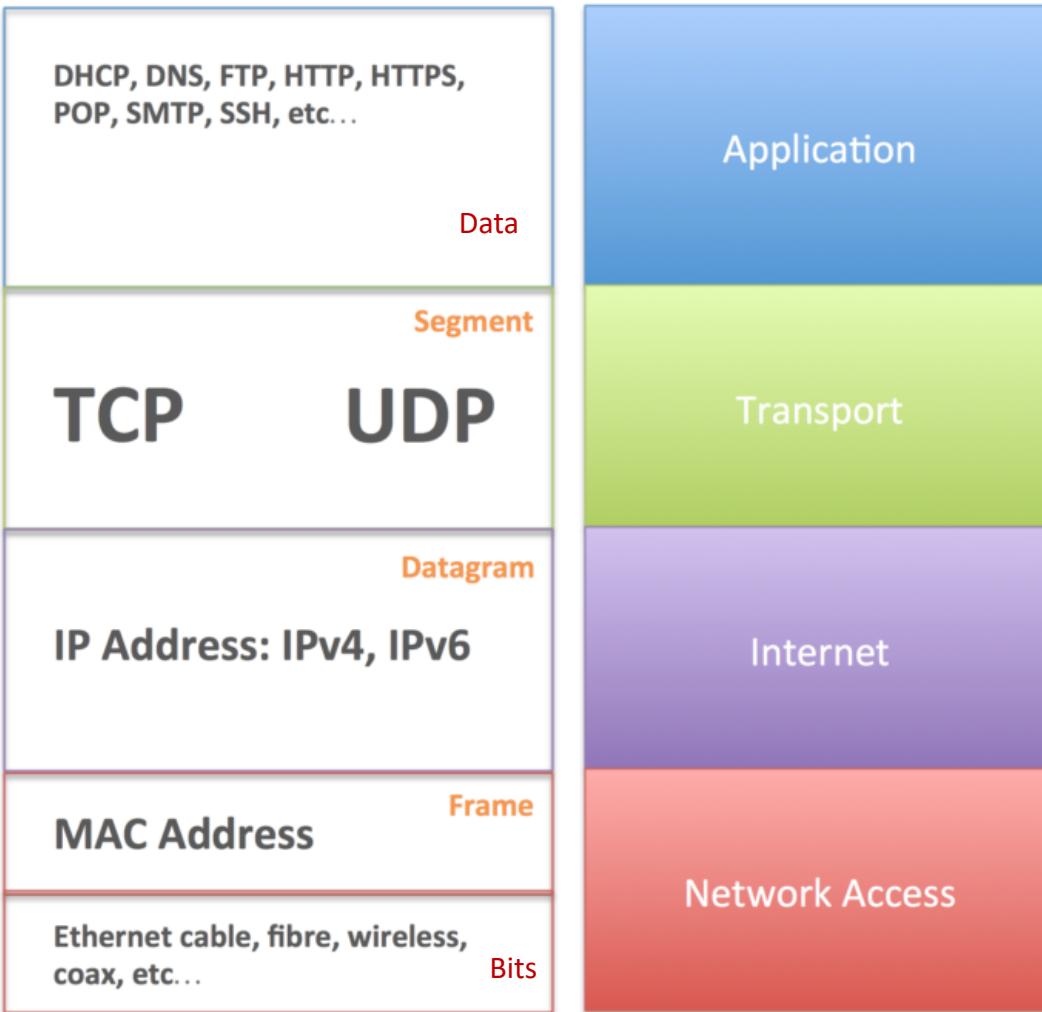


Unit of Communication
– message

The OSI Model

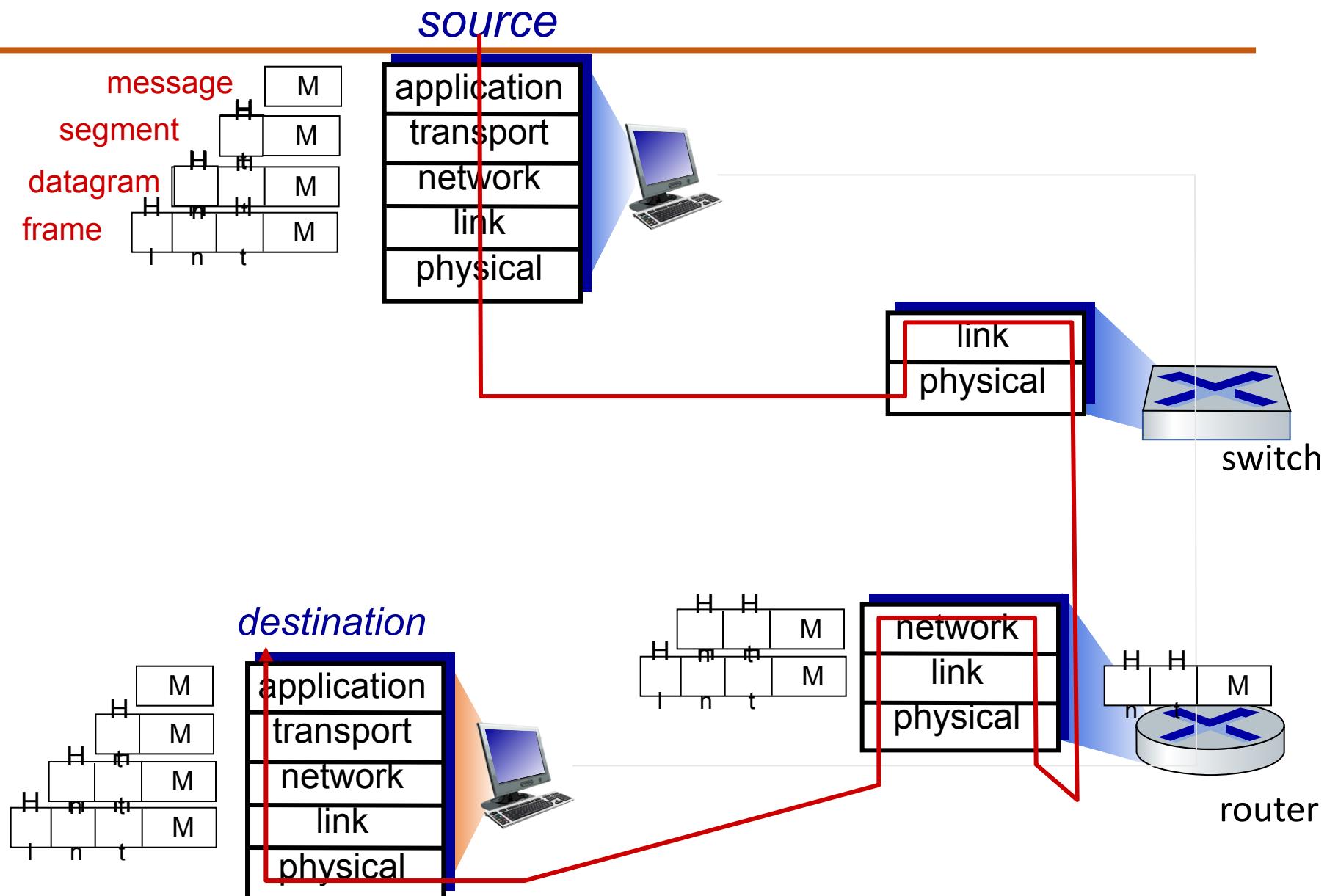


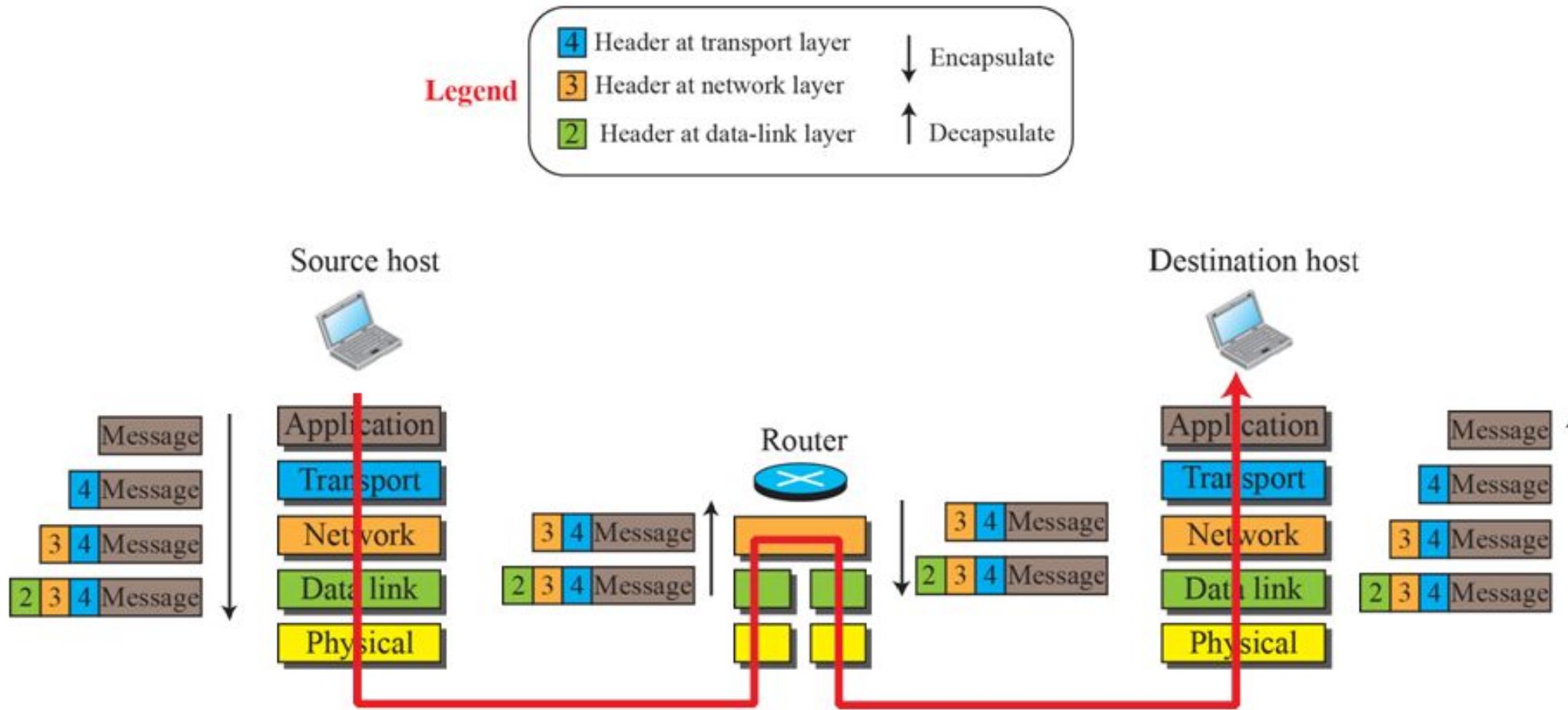
The TCP/IP Model



COMPUTER NETWORKS

Encapsulation – Data Communication in Protocol Stack





Computer Networks and the Internet

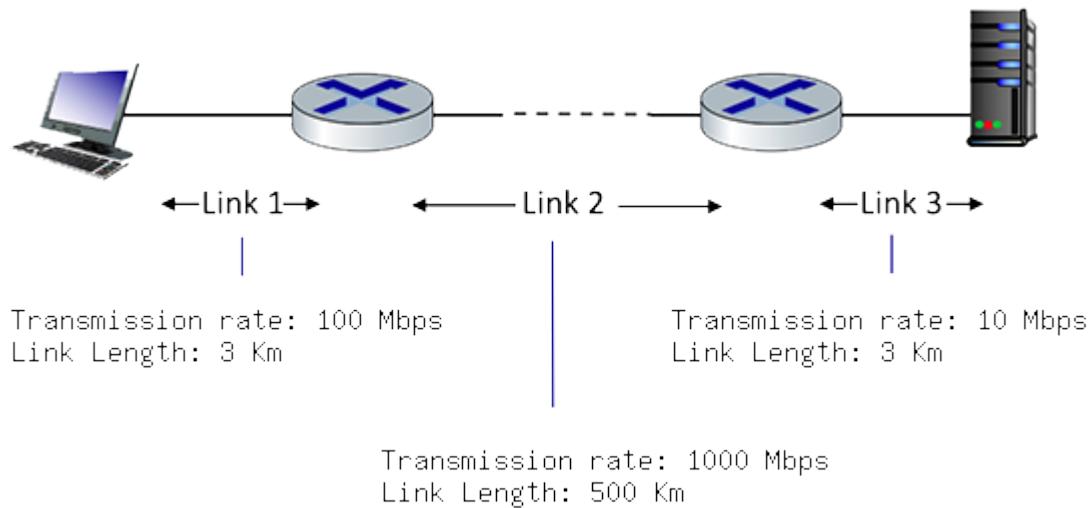
Interactive Exercises

Team Networks

Department of Computer Science and Engineering

Computing end-end delay (transmission and propagation delay)

Consider the figure below, with three links, each with the specified transmission rate and link length.



Find the end-to-end delay (including the transmission delays and propagation delays on each of the three links, but ignoring queueing delays and processing delays) from when the left host begins transmitting the first bit of a packet to the time when the last bit of that packet is received at the server at the right.

The speed of light propagation delay on each link is 3×10^{8} m/sec.

Note that the transmission rates are in Mbps and the link distances are in Km.

Assume a packet length of 8000 bits. Give your answer in milliseconds.

Link 1 transmission delay = $L/R = 8000 \text{ bits} / 100 \text{ Mbps} =$

Link 1 propagation delay = $d/s = 3 \text{ Km} / 3*10^{**8} \text{ m/sec} =$

Link 1 total delay =

Link 2 transmission delay = $L/R = 8000 \text{ bits} / 1000 \text{ Mbps} =$

Link 2 propagation delay = $d/s = 500 \text{ Km} / 3*10^{**8} \text{ m/sec} =$

Link 2 total delay =

Link 3 transmission delay = $L/R = 8000 \text{ bits} / 10 \text{ Mbps} =$

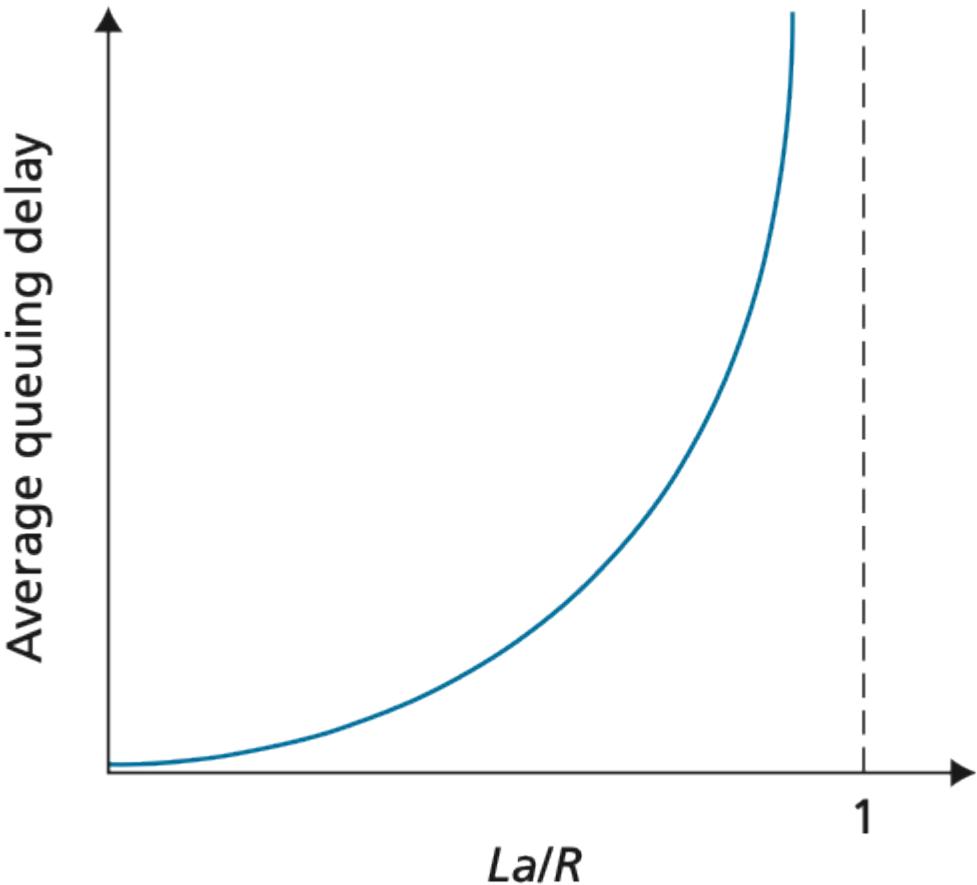
Link 3 propagation delay = $d/s = 3 \text{ Km} / 3*10^{**8} \text{ m/sec} =$

Link 3 total delay =

Total end-to-end delay is the sum of these six delays: **0.0026 msec**.

Computing Queuing Delay

Consider the queuing delay in a router buffer.



Assume:

a constant transmission rate of

$$R = 1800000 \text{ bps},$$

a constant packet-length $L = 6700$ bits, and

a is the average rate of packets/second.

Traffic intensity $I = La/R$, and

Queuing delay is calculated as:

$$I(L/R)(1 - I) \text{ for } I < 1.$$

1. In practice, does the queuing delay tend to vary a lot? Answer with Yes or No

2. Assuming that $a = 30$, what is the queuing delay?

3. Assuming that $a = 76$, what is the queuing delay?

4. Assuming the router's buffer is infinite, the queuing delay is 0.8647 ms, and 1762 packets arrive. How many packets will be in the buffer 1 second later?

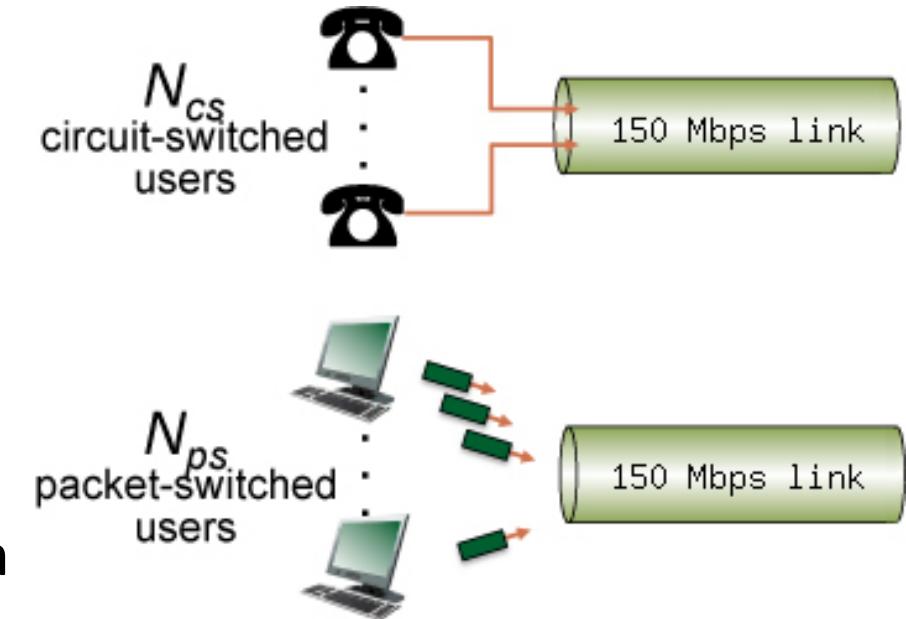
5. If the buffer has a maximum size of 956 packets, how many of the packets would be dropped upon arrival from the previous question?

1. Yes, in practice, queuing delay can vary significantly. We use the above formulas as a way to give a rough estimate, but in a real-life scenario it is much more complicated.
2. Queuing Delay = $I(L/R)(1 - I) * 1000 = 0.1217 * (7300/1800000) * (1-0.1217) * 1000 = 0.4335 \text{ ms.}$
3. Queuing Delay = $I(L/R)(1 - I) * 1000 = 0.3082 * (7300/1800000) * (1-0.3082) * 1000 = 0.8647 \text{ ms.}$
4. Packets left in buffer = $a - \text{floor}(1000/\text{delay}) = 1762 - \text{floor}(1000/0.8647) = 606$ packets.
5. Packets dropped = packets - buffer size = $1762 - 956 = 806$ dropped packets.

Quantitative Comparison of Packet Switching and Circuit Switching

Consider the two scenarios below:

- A circuit-switching scenario in which N_{cs} users, each requiring a bandwidth of 10 Mbps, must share a link of capacity 150 Mbps.
- A packet-switching scenario with N_{ps} users sharing a 150 Mbps link, where each user again requires 10 Mbps when transmitting, but only needs to transmit 30 percent of the time.



Round your answer to two decimals after leading zeros

Interactive Exercise – 3 (Questions)

Answer the following questions:

1. When circuit switching is used, what is the maximum number of circuit-switched users that can be supported? Explain your answer.
2. For rest of the questions, suppose packet switching is used. Suppose there are 29 packet-switching users (i.e., $N_{ps} = 29$). Can this many users be supported under circuit-switching? Explain.
3. What is the probability that a given (*specific*) user is transmitting, and the remaining users are not transmitting?
4. What is the probability that one user (*any one among the 29 users*) is transmitting, and the remaining users are not transmitting? When one user is transmitting, what fraction of the link capacity will be used by this user?
5. What is the probability that any 15 users (of the total 29 users) are transmitting and the remaining users are not transmitting? (Hint: you will need to use the binomial distribution [[1](#), [2](#)]).
6. What is the probability that *more* than 15 users are transmitting? Comment on what this implies about the number of users supportable under circuit switching and packet switching.

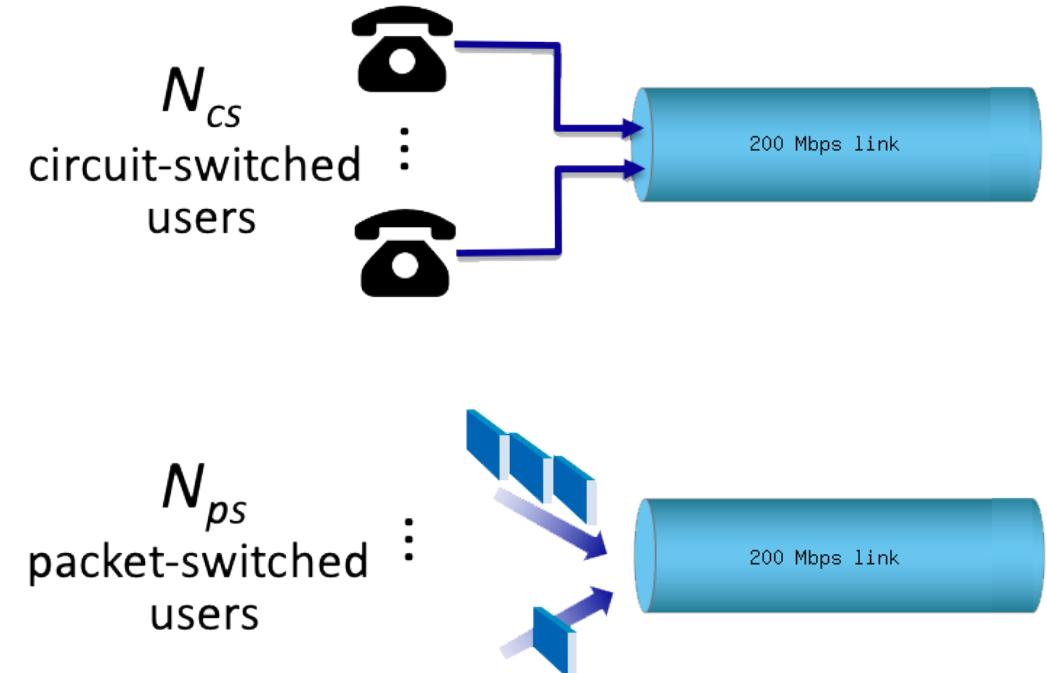
Answers:

1. 15
2. No.
3. 1.37
4. 0.0004
5. 0.00754
6. 0.011653

Quantitative Comparison of Packet Switching and Circuit Switching

Consider the two scenarios below:

- ❑ A circuit-switching scenario in which N_{cs} users, each requiring a bandwidth of 25 Mbps, must share a link of capacity 200 Mbps.
- ❑ A packet-switching scenario with N_{ps} users sharing a 150 Mbps link, where each user again requires 25 Mbps when transmitting, but only needs to transmit 20 percent of the time.



Interactive Exercise – 4 (Questions)

Answer the following questions:

1. When circuit switching is used, what is the maximum number of users that can be supported?
2. Suppose packet switching is used. If there are 15 packet-switching users, can this many users be supported under circuit-switching? Yes or No.
3. Suppose packet switching is used. What is the probability that a given (specific) user is transmitting, and the remaining users are not transmitting?
4. Suppose packet switching is used. What is the probability that one user (*any* one among the 15 users) is transmitting, and the remaining users are not transmitting?
5. When one user is transmitting, what fraction of the link capacity will be used by this user? Write your answer as a decimal.
6. What is the probability that any 10 users (of the total 15 users) are transmitting and the remaining users are not transmitting?
7. What is the probability that *more* than 8 users are transmitting?

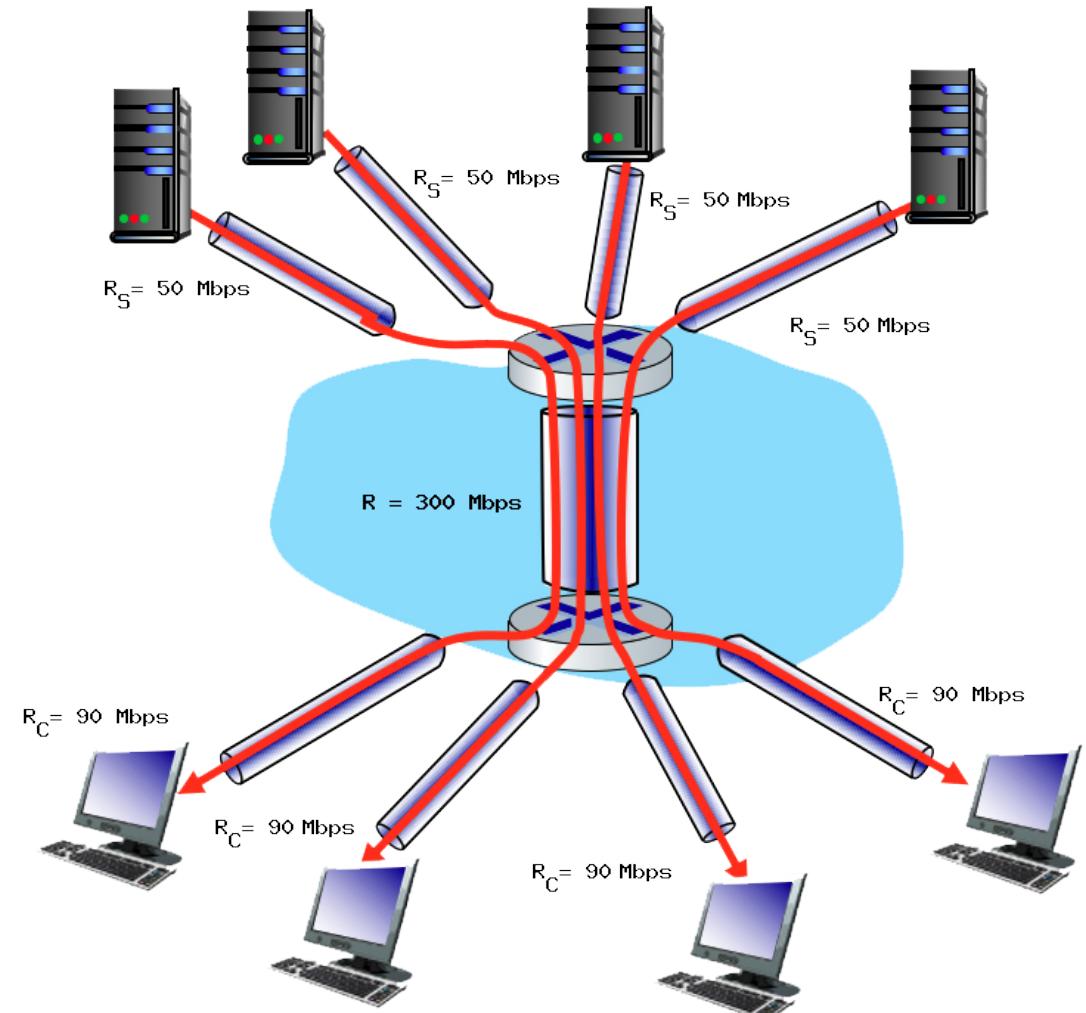
END TO END THROUGHPUT AND BOTTLENECK LINKS

Consider the scenario shown below, with four different servers connected to four different clients over four three-hop paths.

The four pairs share a common middle hop with a transmission capacity of $R = 300$ Mbps.

The four links from the servers to the shared link have a transmission capacity of $R_S = 50$ Mbps.

Each of the four links from the shared middle link to a client has a transmission capacity of $R_C = 90$ Mbps.



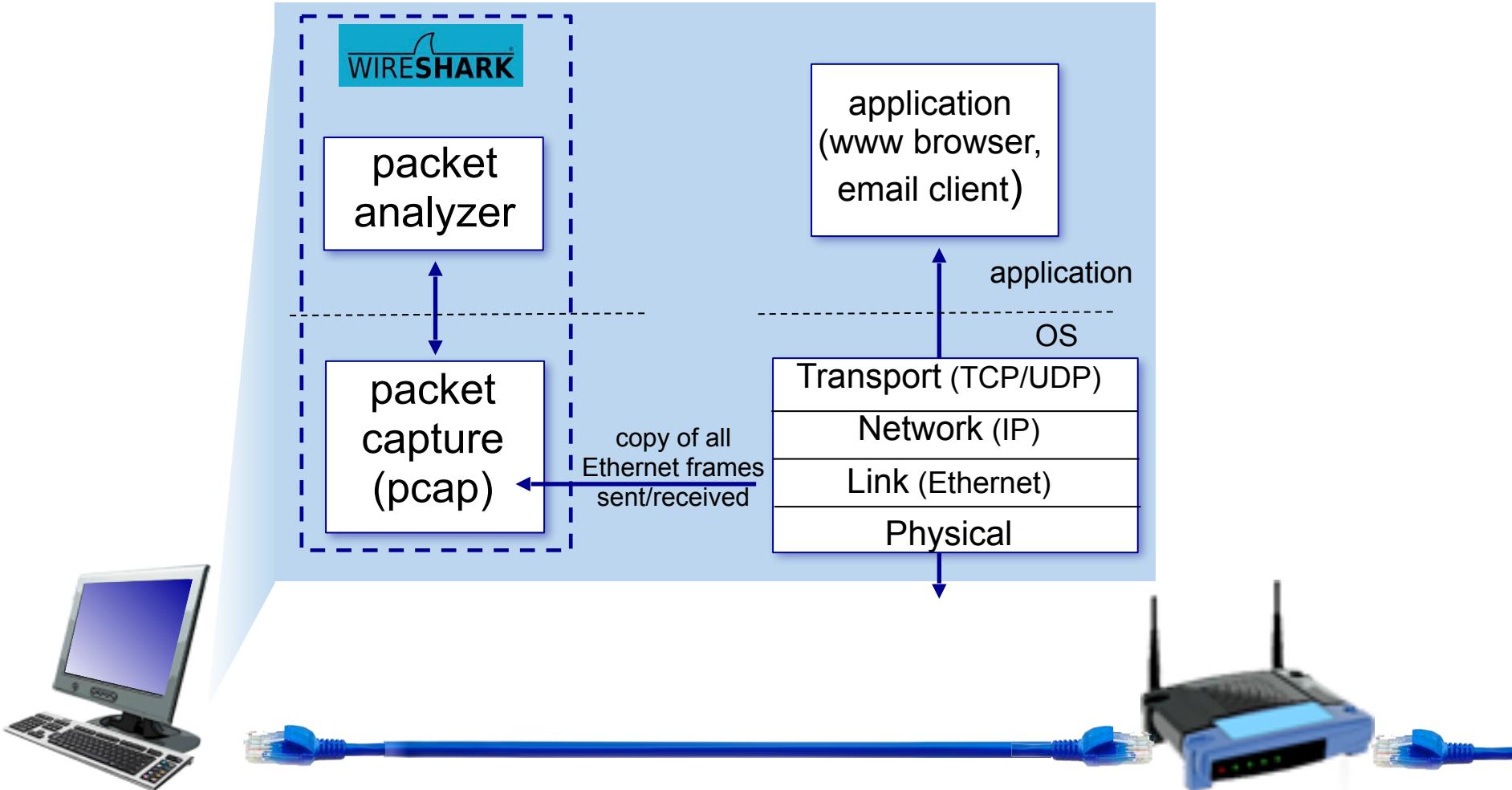
Interactive Exercise – 5 (Questions)

Answer the following questions:

1. What is the maximum achievable end-end throughput (in Mbps) for each of four client-to-server pairs, assuming that the middle link is fairly shared (divides its transmission rate equally)?
2. Which link is the bottleneck link? Format as R_c , R_s , or R
3. Assuming that the servers are sending at the maximum rate possible, what are the link utilizations for the server links (R_s)? Answer as a decimal
4. Assuming that the servers are sending at the maximum rate possible, what are the link utilizations for the client links (R_c)? Answer as a decimal
5. Assuming that the servers are sending at the maximum rate possible, what is the link utilizations for the shared link (R)? Answer as a decimal

Solutions:

1. The maximum achievable end-end throughput is the capacity of the link with the minimum capacity, which is 50 Mbps
2. The bottleneck link is the link with the smallest capacity between R_S , R_C , and $R/4$. The bottleneck link is R_S .
3. The server's utilization = $R_{\text{bottleneck}} / R_S = 50 / 50 = 1$
4. The client's utilization = $R_{\text{bottleneck}} / R_C = 50 / 90 = 0.56$
5. The shared link's utilization = $R_{\text{bottleneck}} / (R / 4) = 50 / (300 / 4) = 0.67$





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THANK YOU

Team Networks

Department of Computer Science and Engineering