

# Lecture 2: Internet Structure

COMP 332, Spring 2018

Victoria Manfredi

WESLEYAN  
UNIVERSITY



**Acknowledgements:** materials adapted from Computer Networking: A Top Down Approach 7<sup>th</sup> edition: ©1996-2016, J.F Kurose and K.W. Ross, All Rights Reserved as well as from slides by Abraham Matta at Boston University and some material from Computer Networks by Tannenbaum and Wetherall.

# Today

## 1. Announcements

- help sessions now Tu (Exley 638) as well as Mo (Exley 618)
- please do the reading!
- to run Python 3: type `python3`
- **homework 1 posted**: may want to wait for lecture 3 for problem 1

## 2. Recap

- direct vs indirect connectivity
- Internet protocol stack

## 3. Internet organization

- **edge**
  - how you connect to Internet
- **core**
  - how your packets get to their destination
  - circuit-switching vs. packet-switching
- **structure**
  - network of networks: internetwork

**Network**

**CONNECTIVITY**

# Connectivity

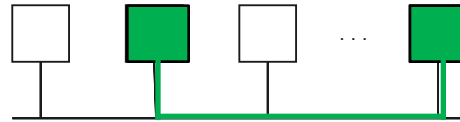
## Direct links

– point-to-point



point-to-point network

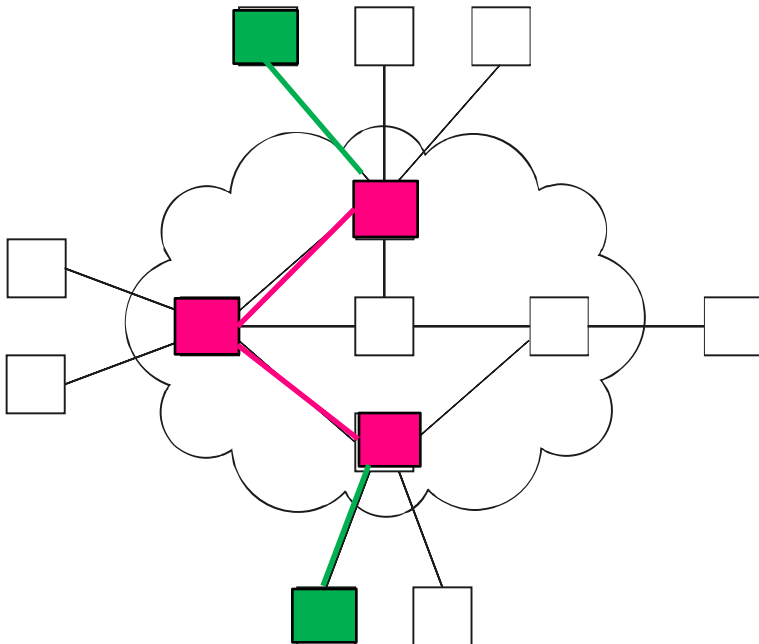
– multiple access



multiple access network

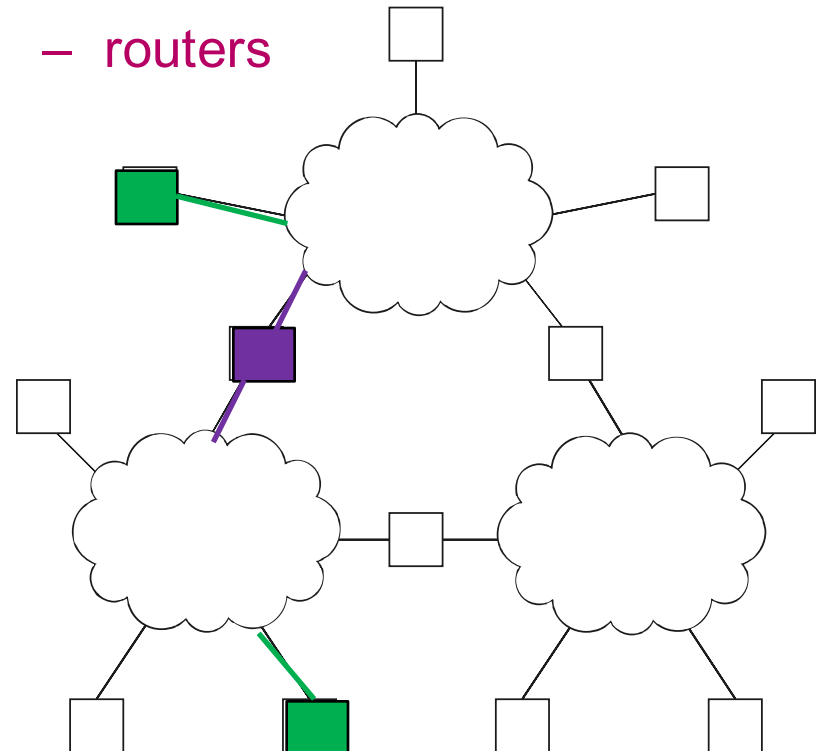
## Indirect connectivity

– switched network



## Internetwork

– routers

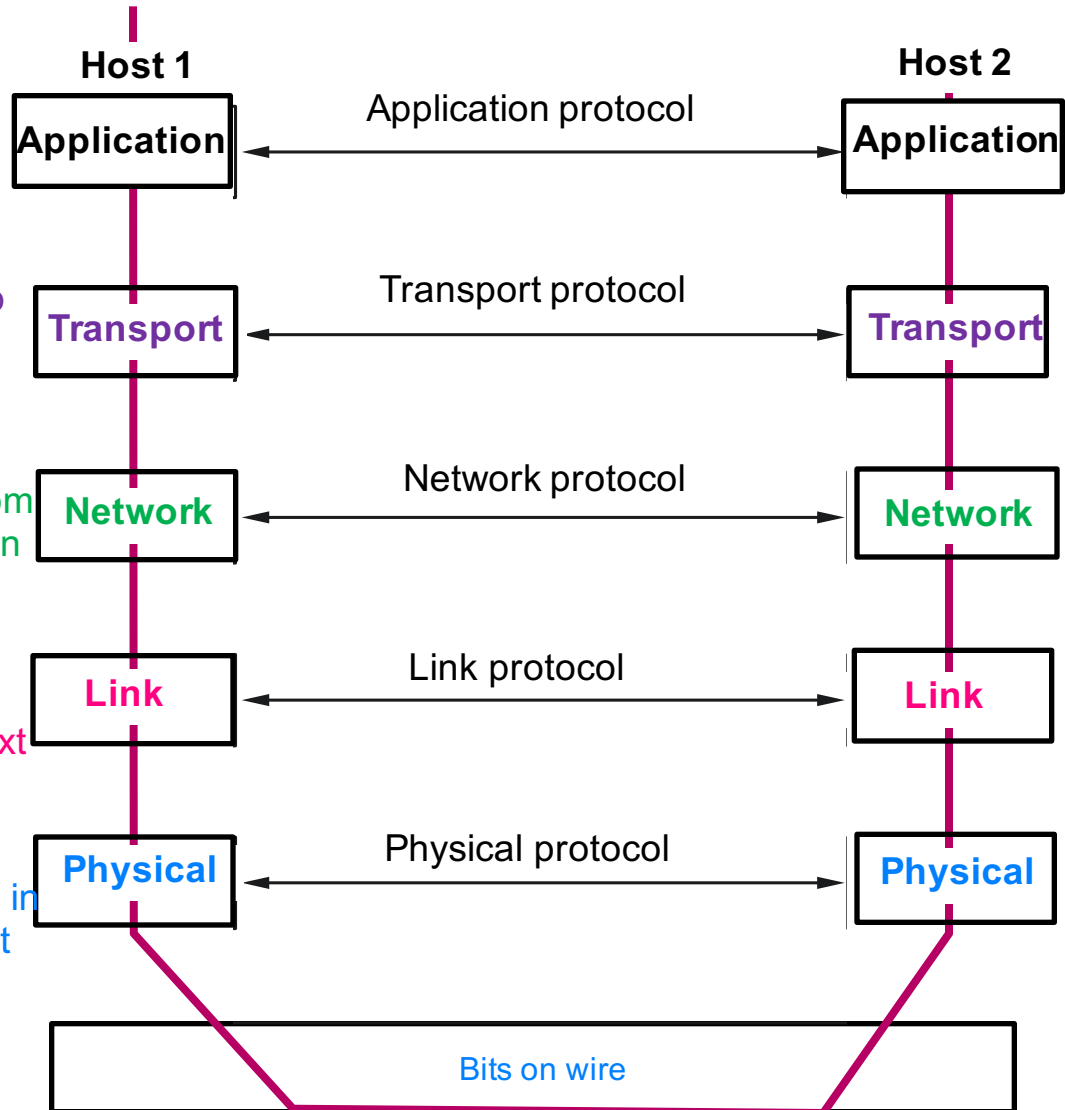


**Internet**

**PROTOCOL STACK**

# Internet protocol stack

Mail service example	Service provided
<ul style="list-style-type: none"> <li>Person reads letter</li> </ul>	<ul style="list-style-type: none"> <li>Support network applications</li> </ul>
<ul style="list-style-type: none"> <li>Letter delivered to person in house</li> </ul>	<ul style="list-style-type: none"> <li>Deliver msgs to app endpoints, flow control, reliability</li> </ul>
<ul style="list-style-type: none"> <li>Letter conveyed between sending house and receiving house</li> </ul>	<ul style="list-style-type: none"> <li>Route segments from source to destination host (indirect)</li> </ul>
<ul style="list-style-type: none"> <li>Letter conveyed from house to mailbox/P.O or from one P.O directly to another P.O.</li> </ul>	<ul style="list-style-type: none"> <li>Move pkt over link from one host to next host (direct)</li> </ul>
<ul style="list-style-type: none"> <li>The actual carrying of the letter across a "link"</li> </ul>	<ul style="list-style-type: none"> <li>Move individual bits in frame from one host to next</li> </ul>



# Internet protocol stack

Layer	Service provided to upper layer	Protocols	Unit of information
5 <b>Application</b>	<ul style="list-style-type: none"><li>Support network applications</li></ul>	FTP, DNS, SMTP, HTTP	<b>Message</b> 1 message may be split into multiple segments
4 <b>Transport</b>	<ul style="list-style-type: none"><li>Deliver messages to app endpoints</li><li>Flow control</li><li>Reliability</li></ul>	TCP (reliable) UDP (best-effort)	<b>Segment</b> (TCP) <b>Datagram</b> (UDP) 1 segment may be split into multiple packets
3 <b>Network</b>	<ul style="list-style-type: none"><li>Route segments from source to destination host</li></ul>	IP (best-effort) Routing protocols	<b>Packet</b> (TCP) <b>Datagram</b> (UDP)
2 <b>Link</b>	<ul style="list-style-type: none"><li>Move packet over link from one host to next host</li></ul>	Ethernet, 802.11	<b>Frame</b> MTU is 1500 bytes
1 <b>Physical</b>	<ul style="list-style-type: none"><li>Move individual bits in frame from one host to next</li><li>“bits on wire”</li></ul>	Ethernet phy 802.11 phy Bluetooth phy DSL	<b>Bit</b>

# Internet protocol stack

## Where to place functionality in Internet?

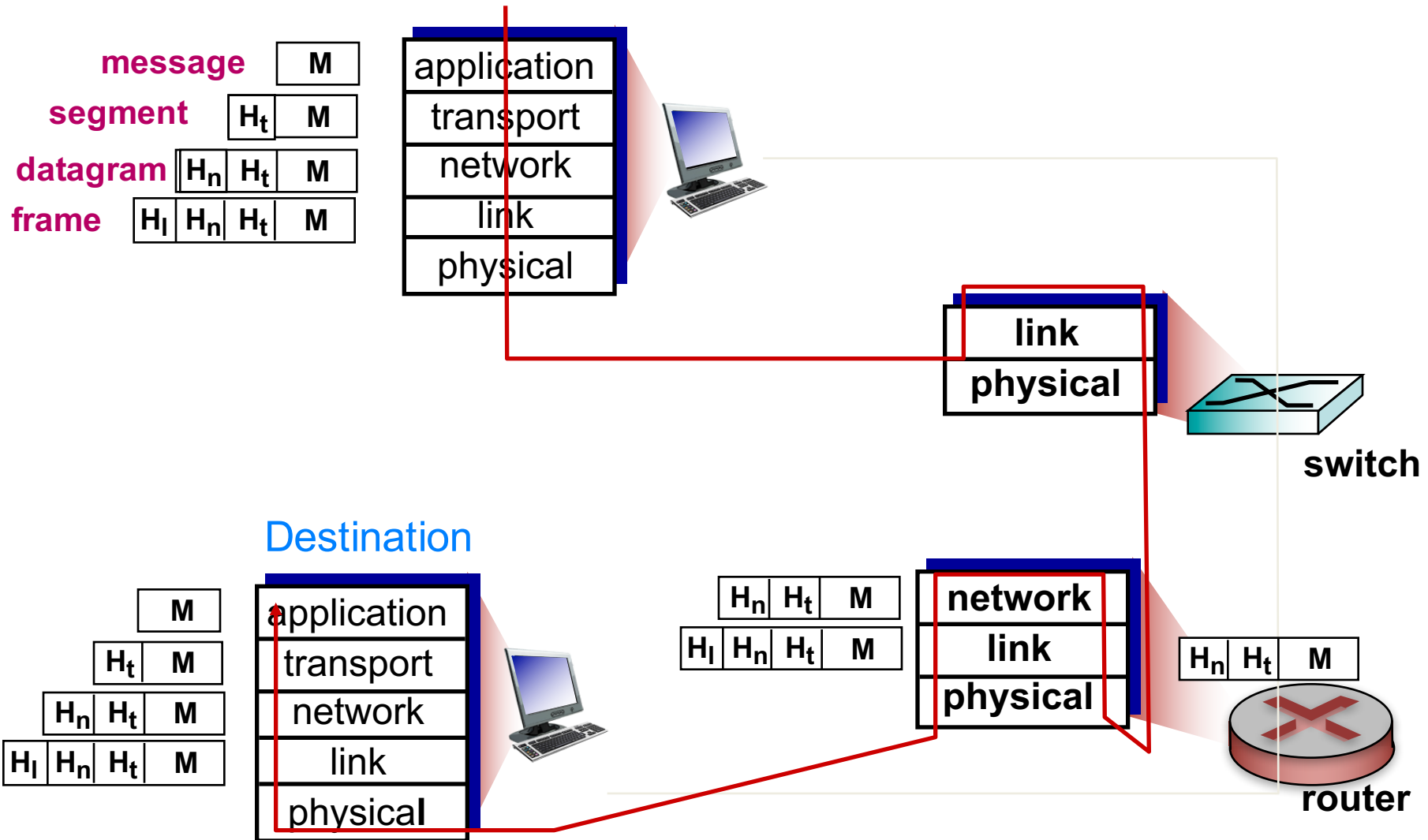
- Option 1:
  - inside network (switches/routers)
- Option 2:
  - at edges (hosts)

## Illustrates “end-end” principle

- some network functionality can only be correctly implemented at end-hosts
- e.g., file transfer
  - should each link check or end hosts check?
  - what if a link on path fails?

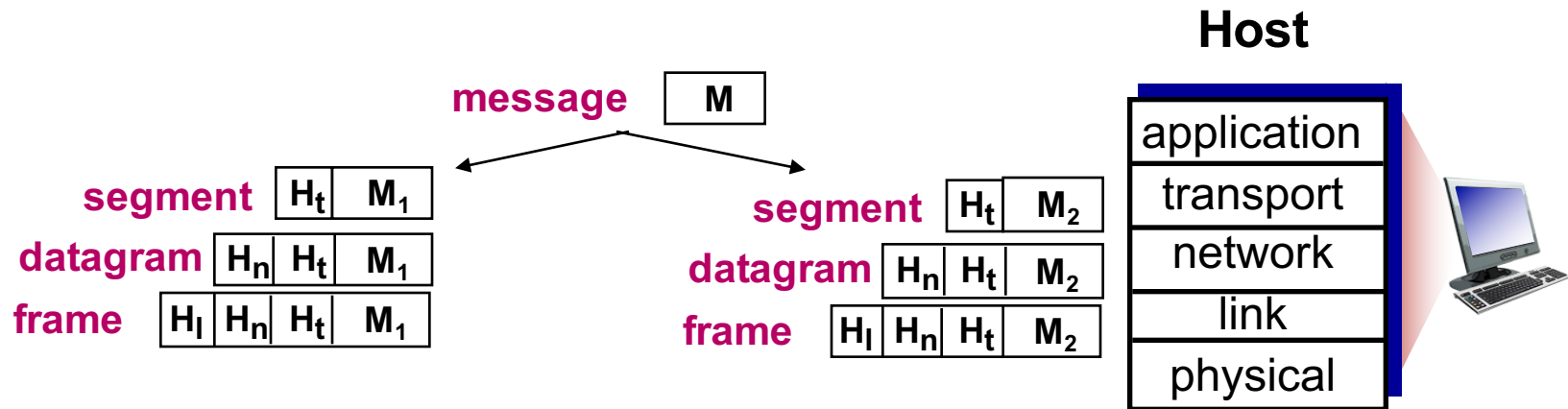


# Encapsulation/Decapsulation



# Fragmentation/Assembly

Why fragment? Max size of Ethernet frame is specified to be 1522 bytes

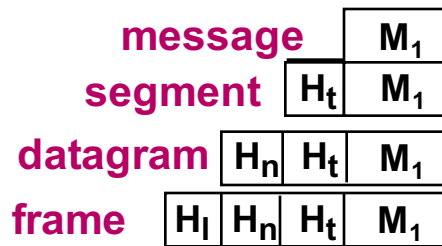


Now some additional book-keeping to keep track of which **segments** belong to which **message**

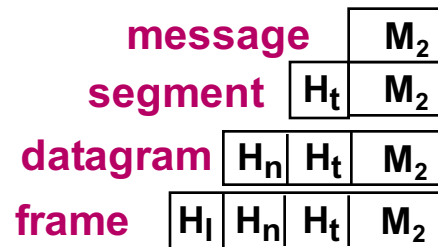
# Multiplexing/Demultiplexing

**Why multiplex?** Many processes sending network traffic simultaneously on host, many hosts sharing network

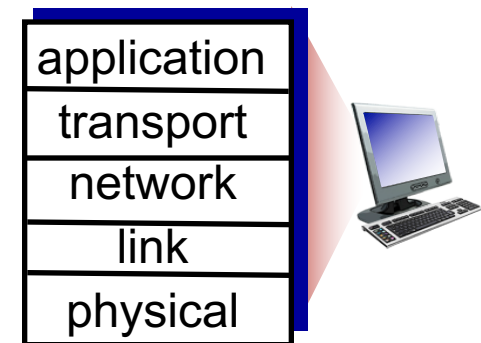
## Process 1: Web



## Process 2: Email



## Host



Now some additional book-keeping to keep track of which **segments** belong to which **process** on host

**Internet**

**COMPONENTS**

# How is Internet organized physically?

## A network of networks: internetwork

- every device implements IP (Internet Protocol) and has IP address

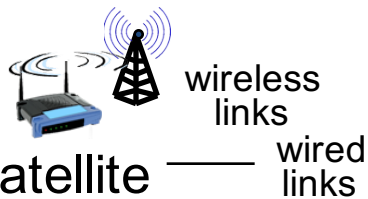
## Billions of connected devices

- run network apps



## Communication links

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

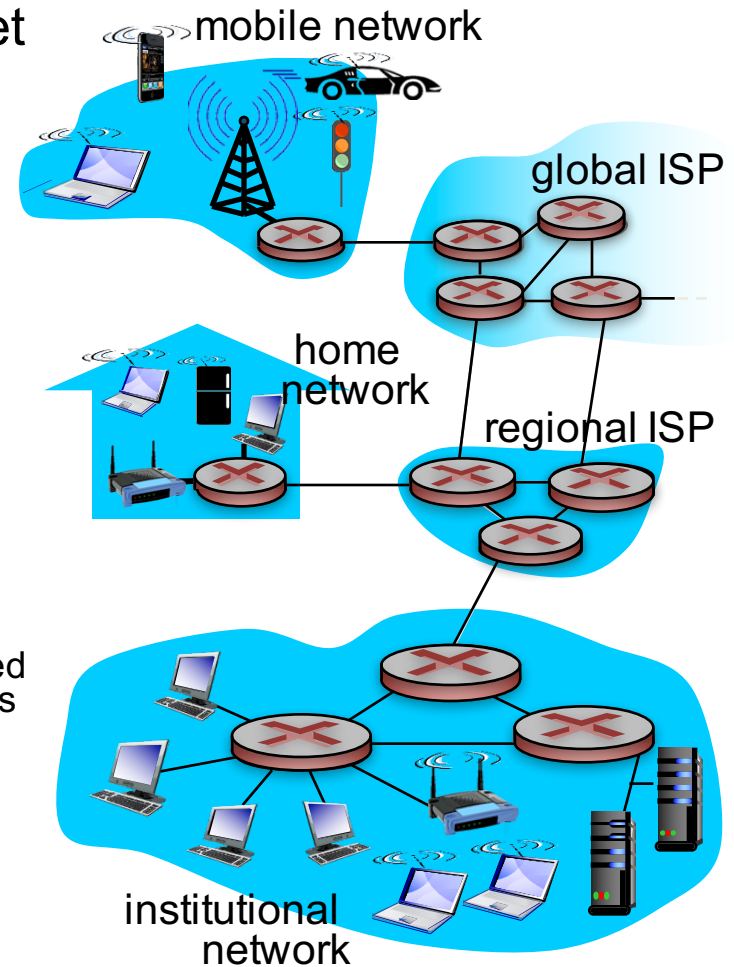


## Routers (and switches)

- forward packets (and frames)

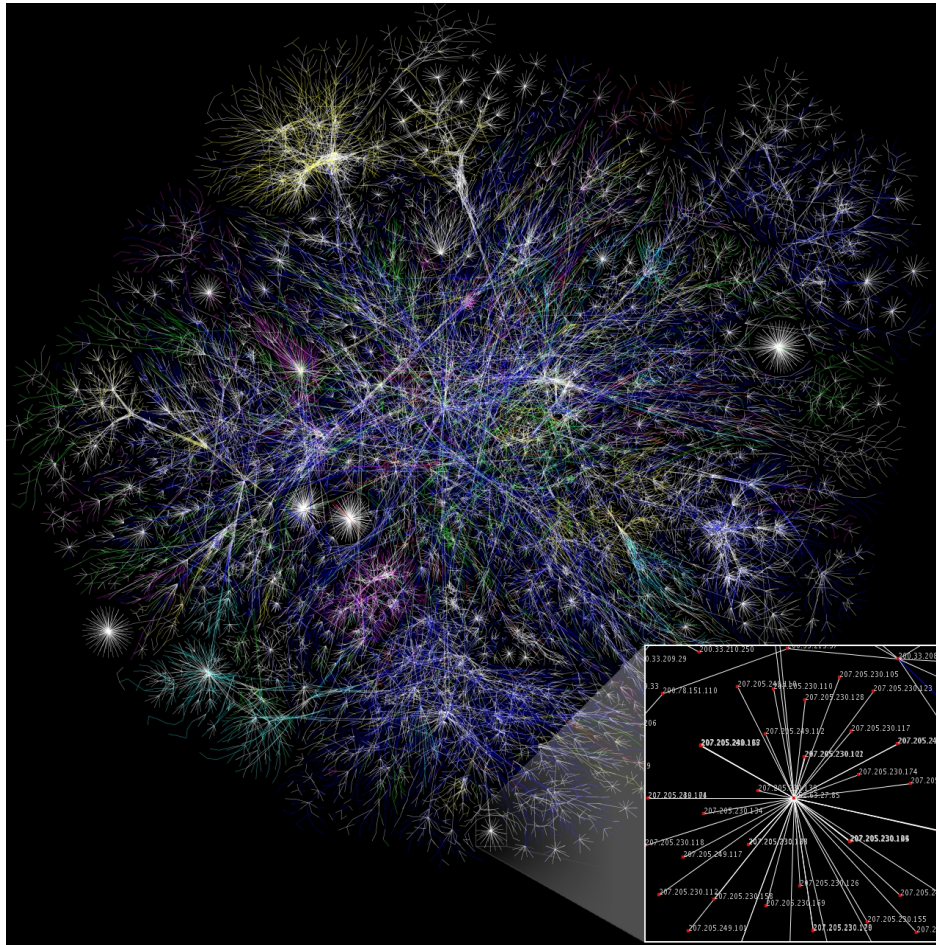


*vumanfredi@wesleyan*



ISP: Internet Service Provider

# Map of the Internet



Who is connected to whom?

## Nodes

- IP addresses of devices

## Edges

- lengths are delay between 2 devices

By The Opte Project [CC BY 2.5  
(<http://creativecommons.org/licenses/by/2.5>)], via  
Wikimedia Commons

# How is Internet structured?

## Network edge

- **hosts**: clients and servers
- servers often in data centers

## Access networks, physical media

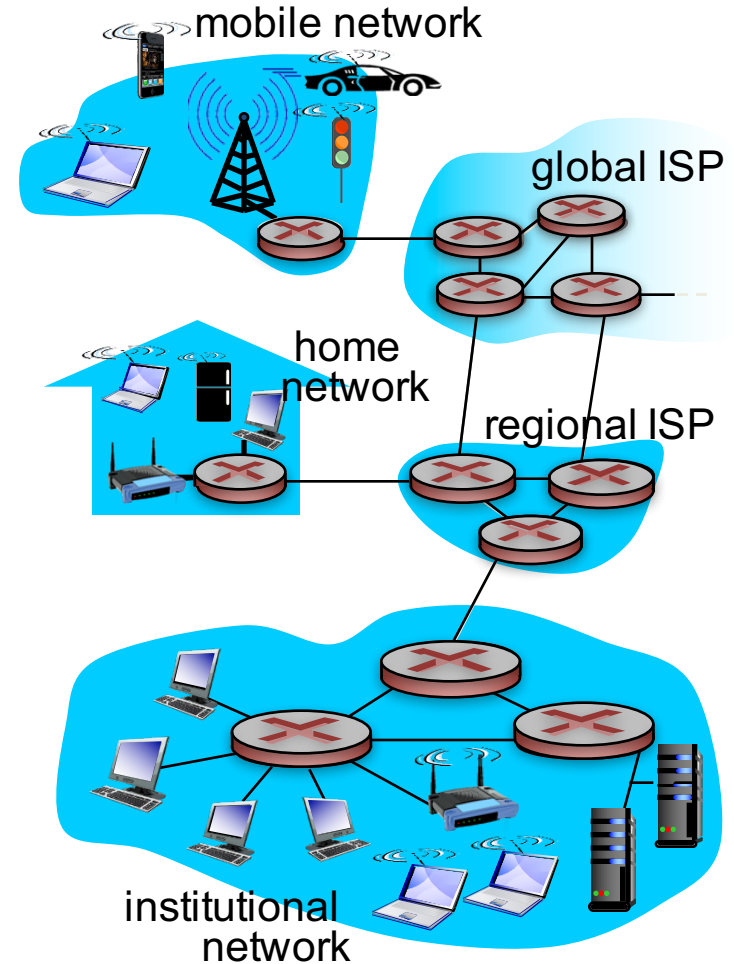
- wired, wireless communication **links**

## Network core

- interconnected **routers**
- network of networks

## Protocols

- control message sending, receiving



ISP: Internet Service Provider

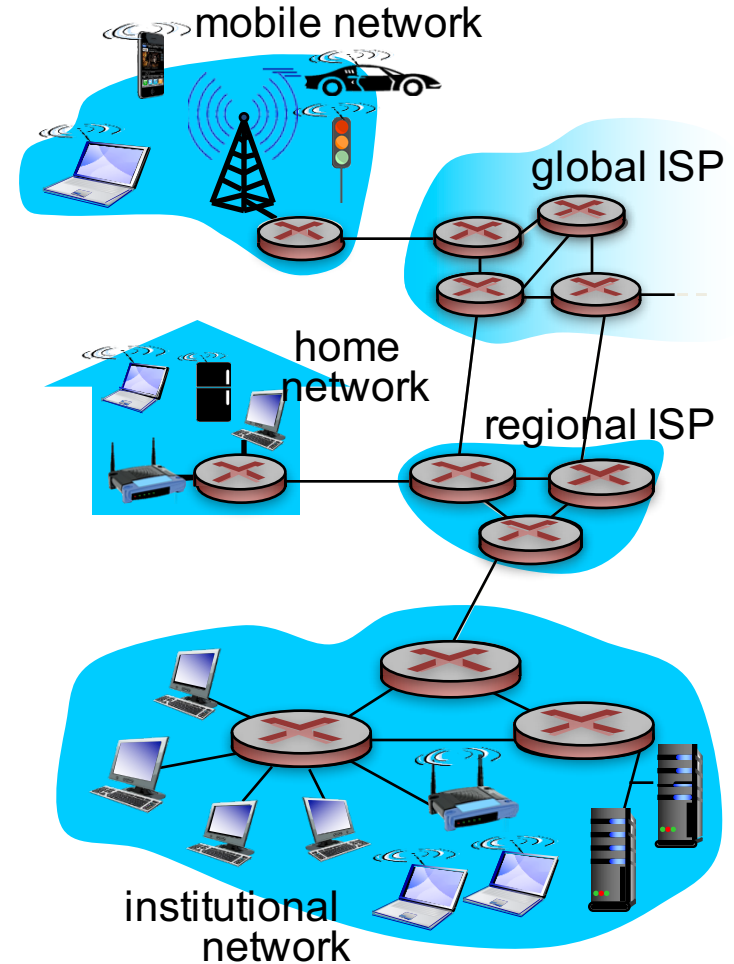
# Internet provides services

## Services to applications

- E.g., web, VoIP, email, games, ecommerce, social nets, ...

## Programming interface to apps

- **hooks**
  - for sending and receiving app programs to connect to Internet
- **service options**
  - analogous to postal service





**Internet**

**EDGE**

# How do you connect to Internet?

## Hosts connect to edge router

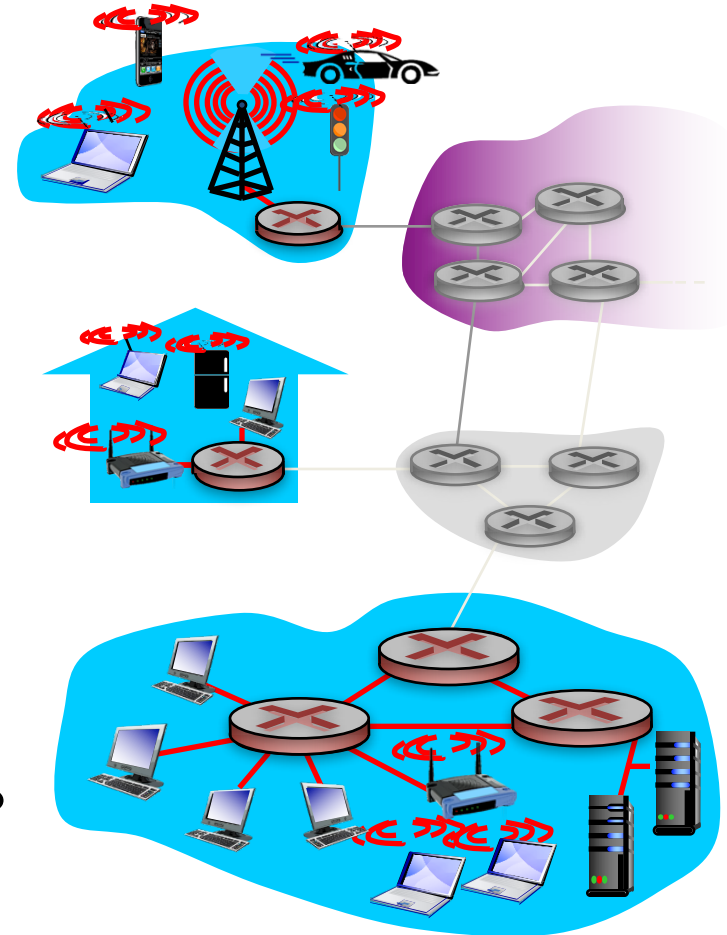
- access network/ISP

## Access networks

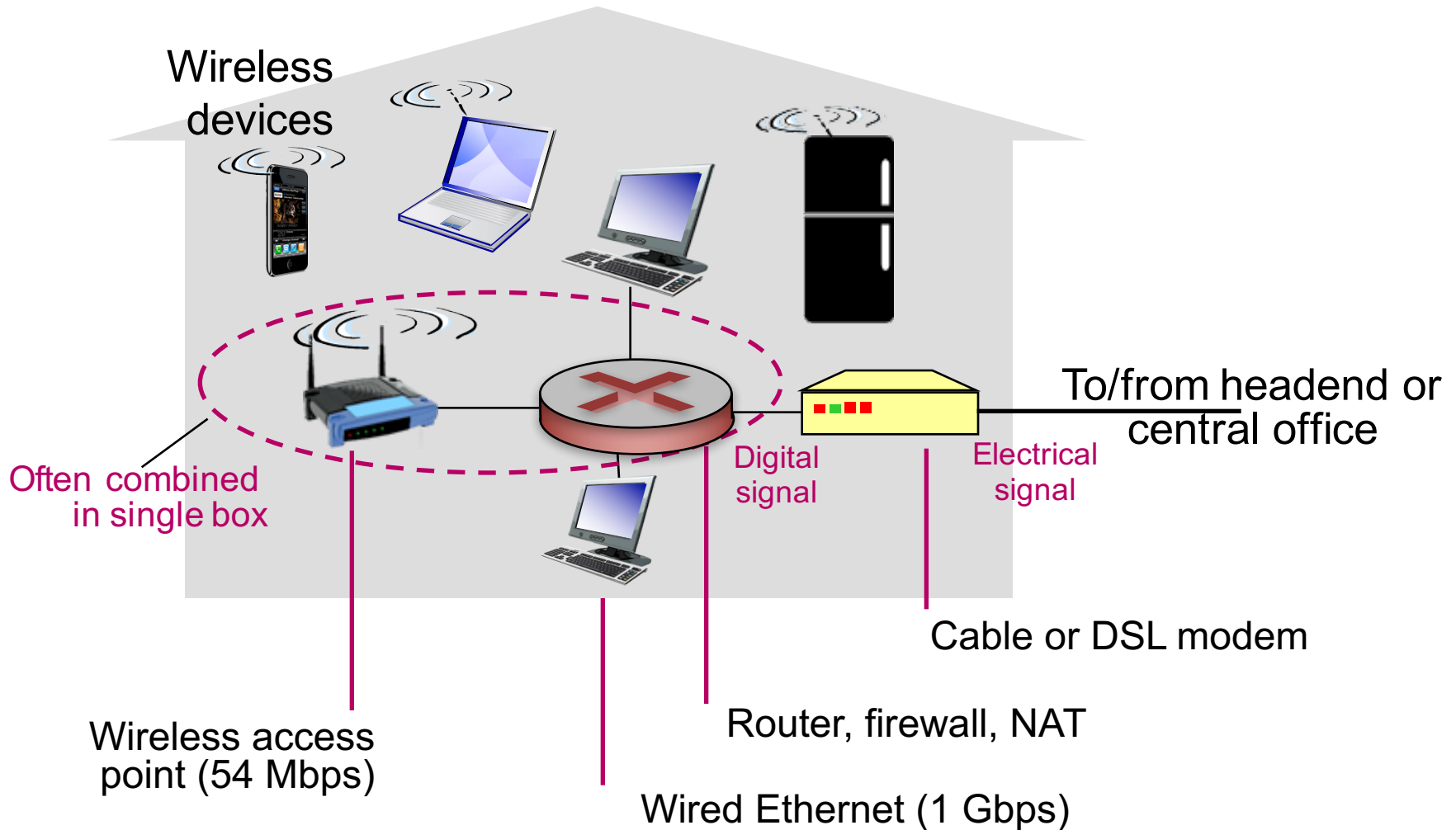
- residential
  - DSL (telephone), cable,
- institutional
  - school, company
- mobile

## Issues

- bandwidth (bps) of access network?
- shared or dedicated?



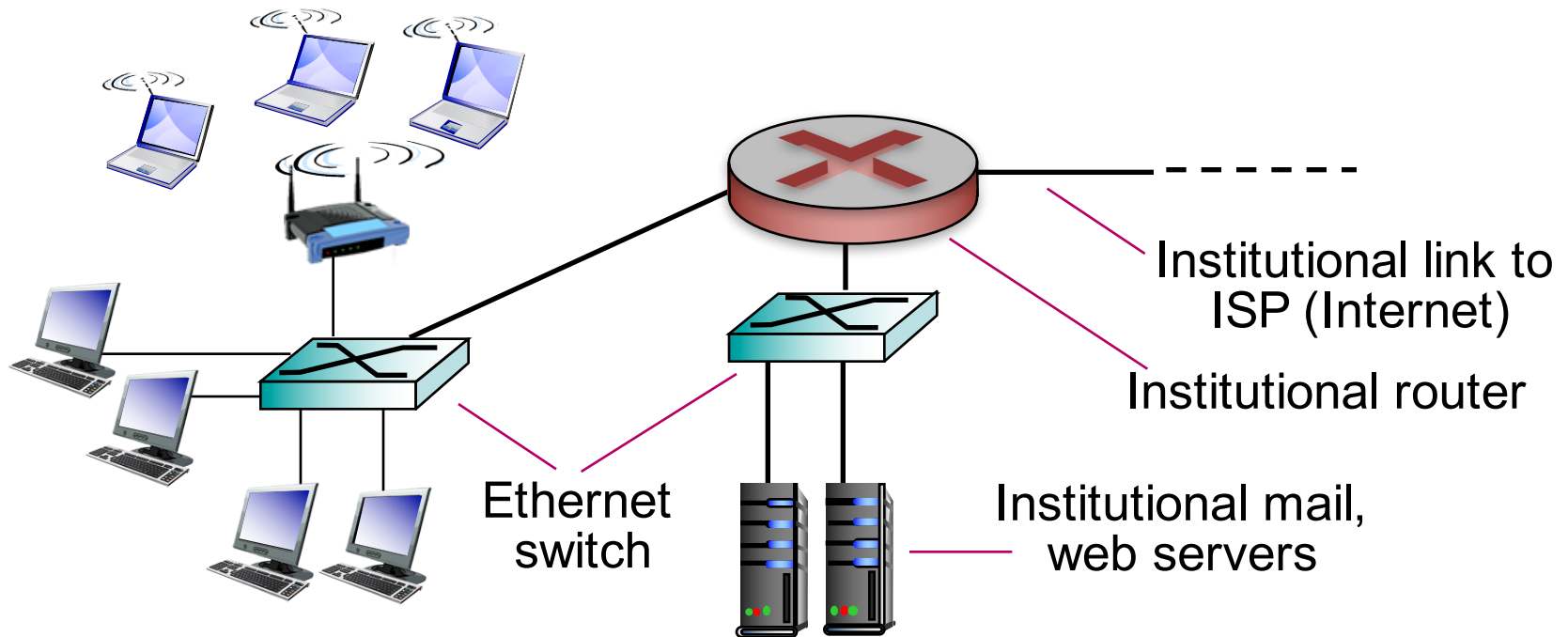
# Access network: home network



# Access network: enterprise (Ethernet)

Typically used in companies, universities, etc.

- 10 Mbps, 100Mbps, 1Gbps, 10Gbps transmission rates
- today, end systems typically connect into Ethernet switch



# Access network: wireless

## Shared wireless access network

- connects end system to router via base station (aka “access point”)

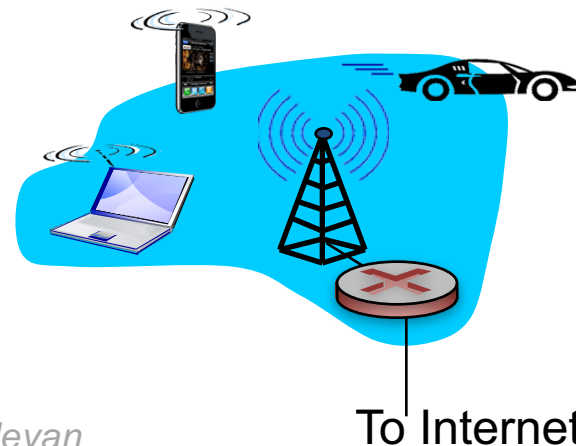
## Wireless LANs

- within building (100 ft.)
- 802.11b/g/n (WiFi):
  - 11, 54, 450 Mbps



## Wide-area wireless access

- provided by telco (cellular) operator, 10's km
- between 1 and 10 Mbps
- 3G, 4G: LTE



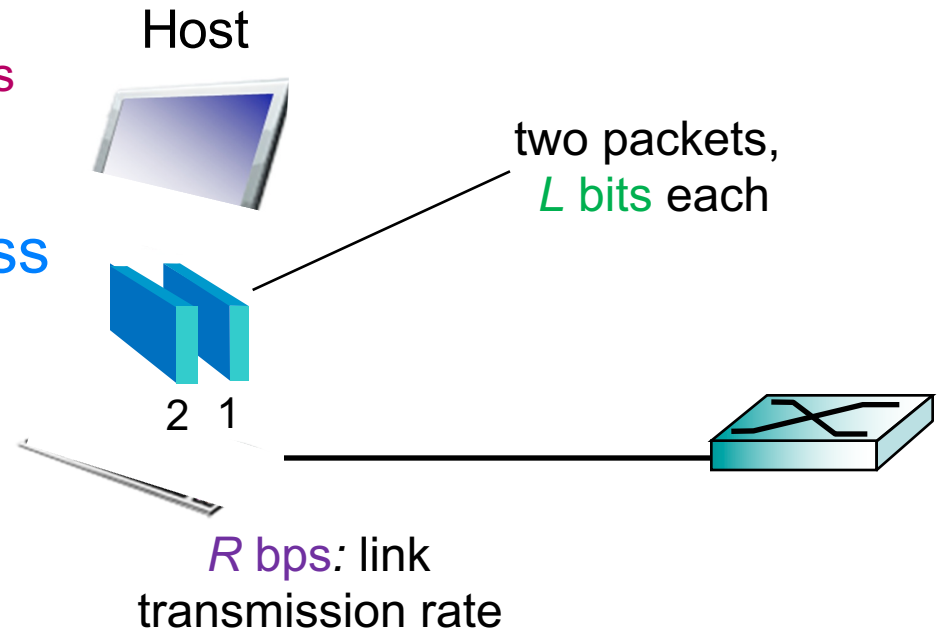
# Host sends packets of data

## 1. Given application message

- breaks into **packets**
  - smaller chunks of length  $L$  bits

## 2. Transmit packets into access network

- at **transmission rate  $R$** 
  - aka link capacity
  - aka link bandwidth



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

**Internet**

**CORE**

# How to move data through Internet core?

## Internet core

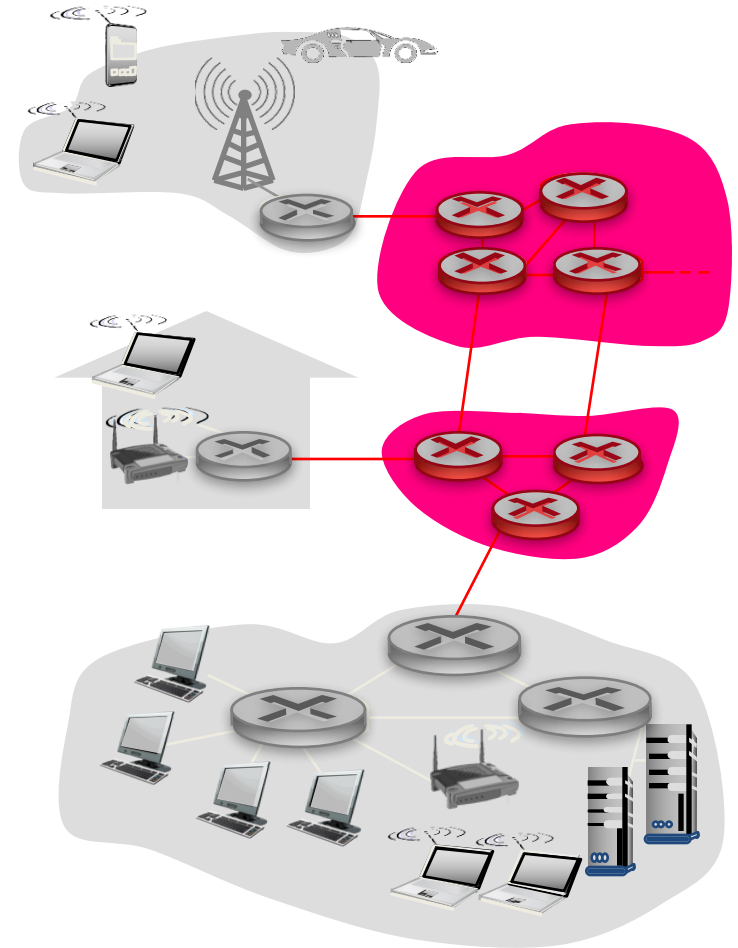
- mesh of interconnected routers

## Option 1: Packet-switching

- on-demand resource allocation
- best effort service
- good bandwidth use

## Option 2: Circuit-switching

- reserved resources
- guaranteed service
- may waste bandwidth

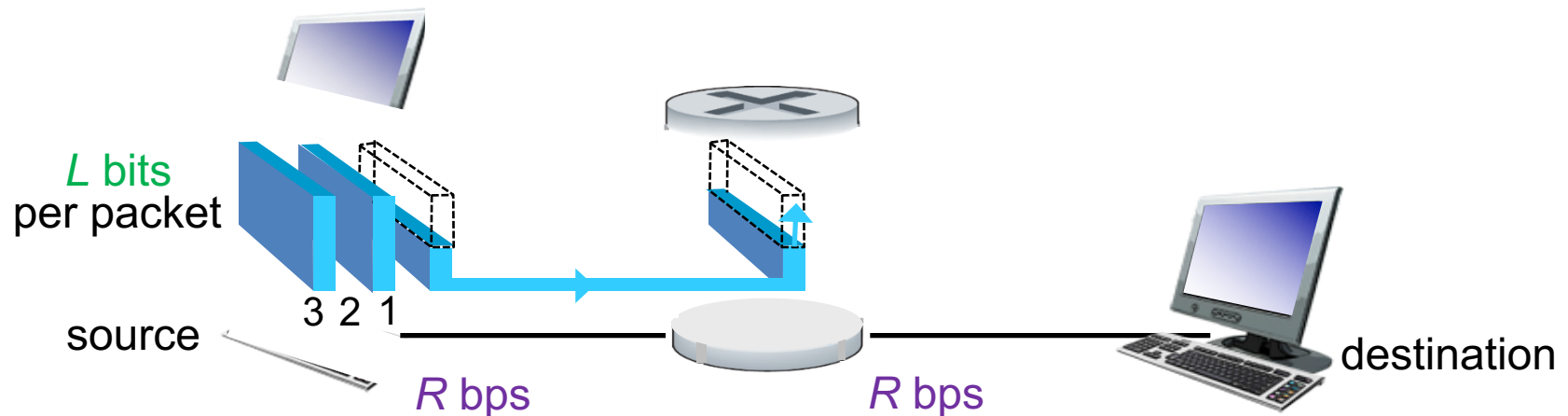


Let's see why packet-switching is used in core



# Packet-switching: store-and-forward

1. Hosts break app-layer messages into packets



2. Time to transmit (push out)  $L$ -bit packet into  $R$  bps link:  
 $L / R$  seconds

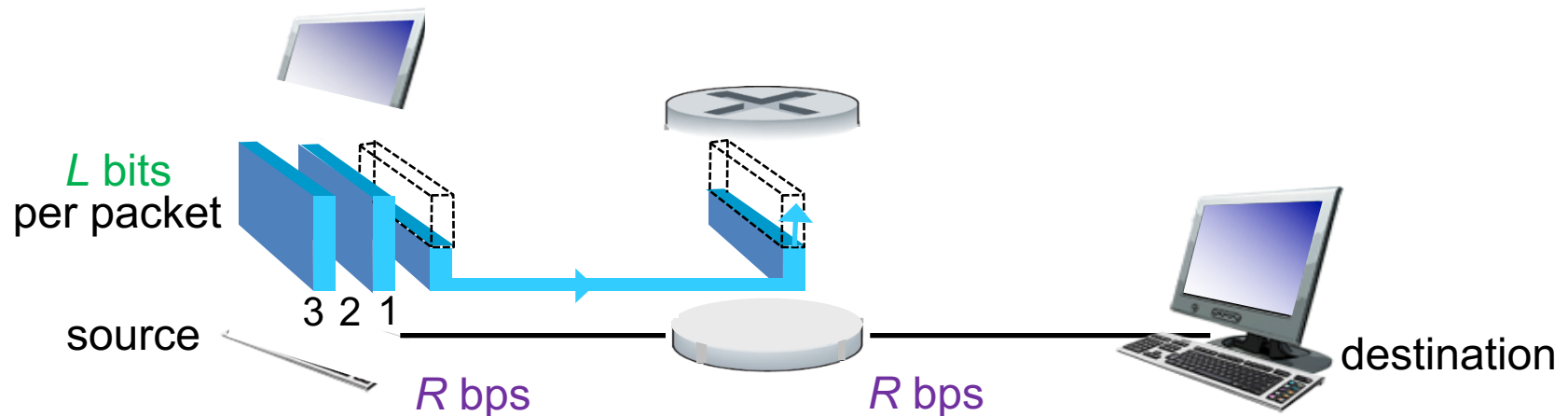
## 1-hop example

- $L = 7.5$  Mbits
- $R = 1.5$  Mbps
- 1-hop transmission delay = 5 sec

# Packet-switching: store-and-forward

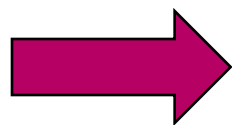
1. Hosts break app-layer messages into packets

3. Store-and-forward: entire packet must arrive at router before it can be transmitted on next link



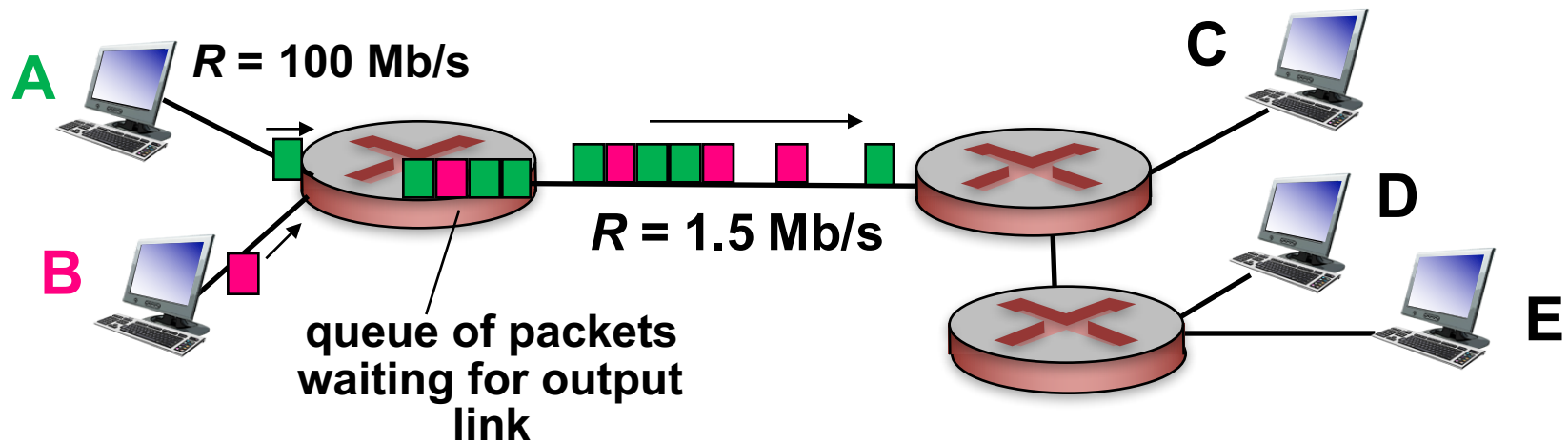
2. Time to transmit (push out)  
 $L$ -bit packet into  $R$  bps link:  
 $L / R$  seconds

4.  $L / R$  seconds



End-end transmission delay =  $2 L / R$   
(assuming zero propagation delay)

# Packet-switching: queueing delay, loss



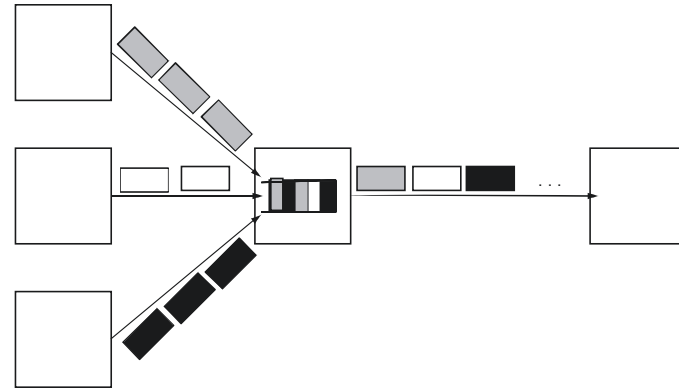
If link arrival rate (in bits)  $>$  transmission rate link for some time

- packets will **queue**, wait to be transmitted on link
- packets can be **dropped** (lost) if memory (buffer) fills up

# Packet-switching: multiplexing users

## Multiplexing

- share links and network resources among multiple users



## Statistical Multiplexing

- time-division, but **on demand** rather than fixed (no waste)
- reschedule link on a per-packet basis
- packets from different sources interleaved on link
- buffer packets that are contending for link
- packet queue may be processed FIFO, but not necessarily
- buffer overflow, causing **packet drop** (loss), is called **congestion**

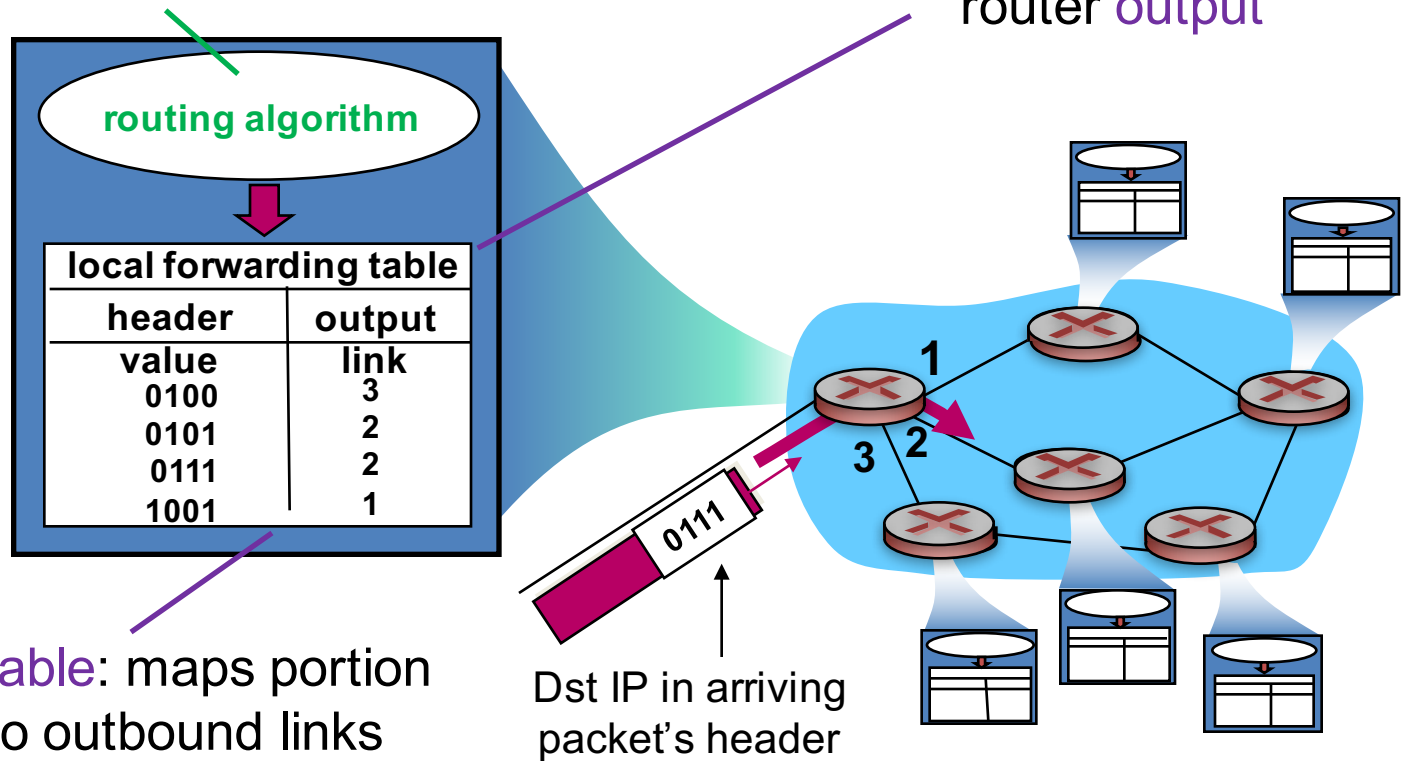
# Packet-switching: 2 key functions of Internet core

How does Internet router determine outgoing link for packet?

– uses **destination IP** address in packet

1. **Routing**: determines src-dst **route** taken by packets, used to set forwarding table

2. **Forwarding**: move pkts from router's **input** to appropriate router **output**



**Forwarding table**: maps portion of dst IPs to outbound links

Dst IP in arriving packet's header

# Alternative core: circuit switching

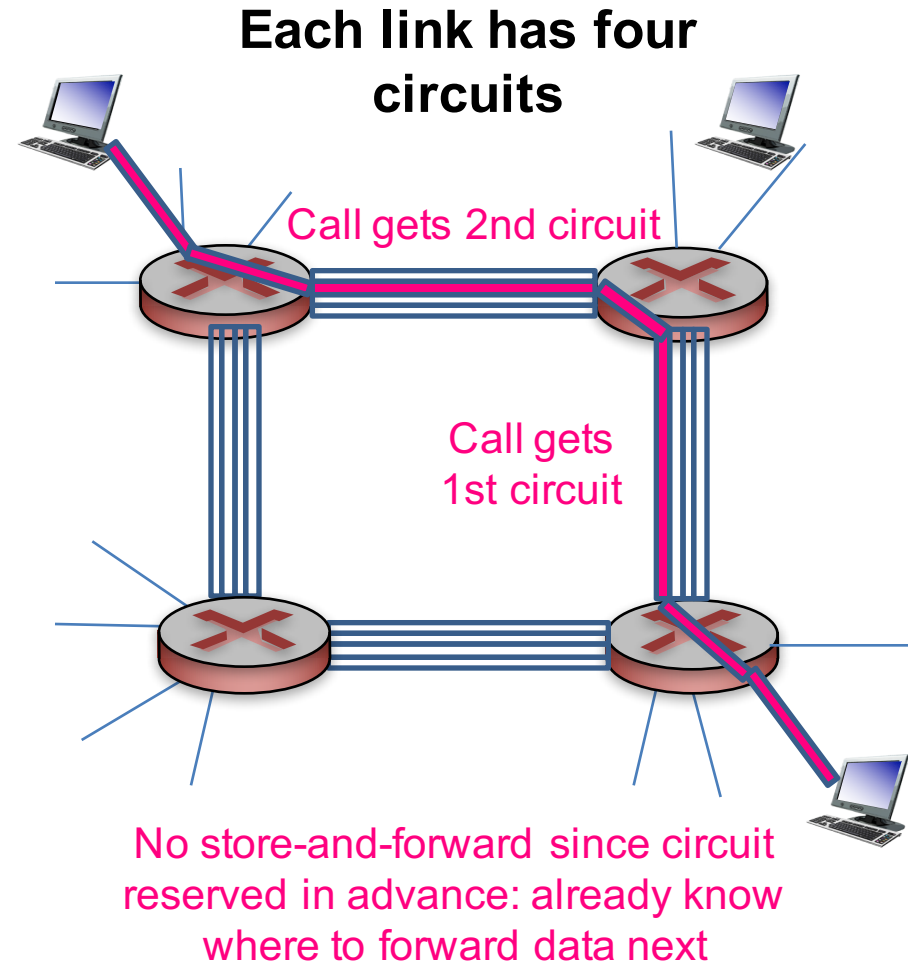
## End-end resources allocated

- reserved for “call” between source & dest

## Dedicated resources

- no sharing
- circuit-like (guaranteed) performance
- circuit segment idle if not used by call (no sharing)

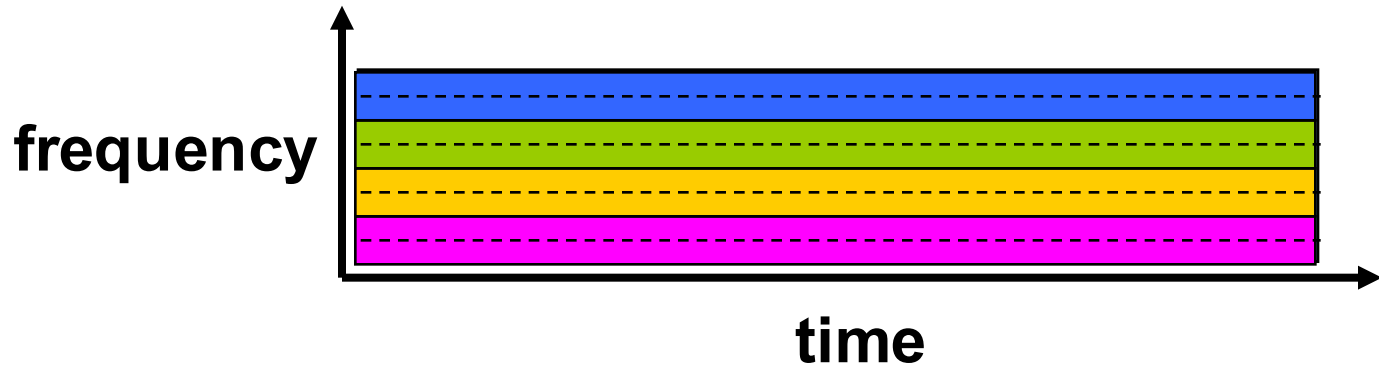
## Commonly used in traditional telephone networks



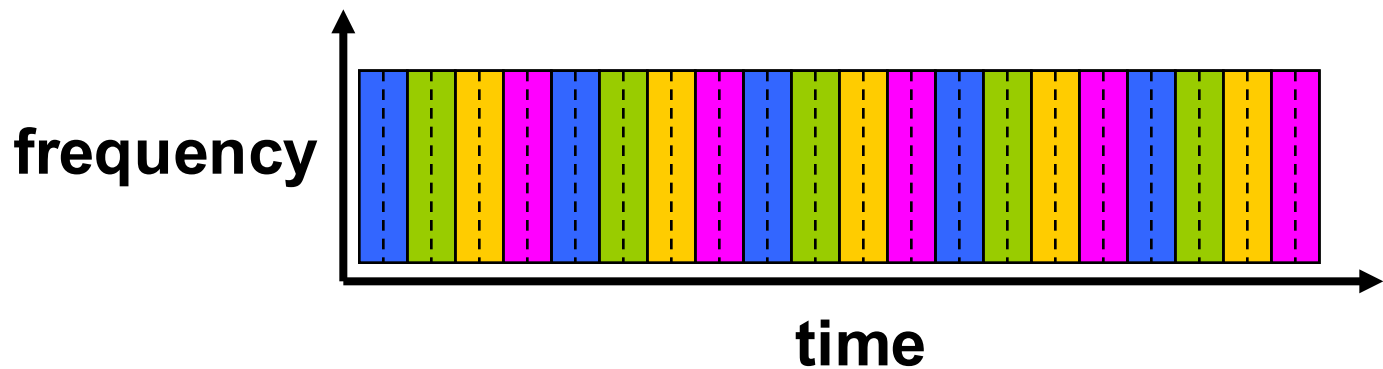
**Q: what happens if there is a lull in conversation?**

# Circuit switching: multiplexing users

## Frequency Division Multiplexing



## Time Division Multiplexing

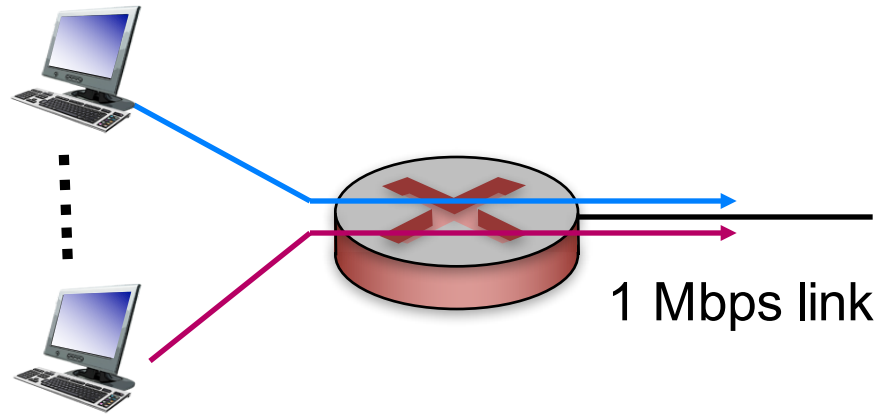


4 users

# Packet switching versus circuit switching

$N$  users

- each user is active 10% of time
- 100 Kbps when active



How many users can be supported?

Circuit switching

- 1 Mbps / 10 = 100 Kbps
- $N = 10$  users

Packet switching

- $N = 35$  users
- prob  $> 10$  users active at same time is  $< .0004$

Q: how did we get value 0.0004?

Q: what happens if  $> 35$  users ?

Packet switching allows more users to use network!



# Binomial random variable (homework)

Suppose we do  $n$  independent experiments, each of which succeeds with probability  $p$  and fails with probability  $1-p$

$X$  = R.V. indicating # of successes that occur in  $n$  trials

$$P(X = i) = \underbrace{\binom{n}{i}}_{\substack{\text{n choose i} \\ \text{different ways to} \\ \text{get i successful} \\ \text{experiments}}} p^i \underbrace{(1-p)^{n-i}}_{\text{failures}}$$

successes

**Independent experiments:**

knowledge about one experiment occurring does not affect probability of other experiment occurring: e.g., coin toss.

$$P(A \text{ and } B) = P(A) \times P(B)$$

$$P(A \text{ or } B) = P(A) + P(B)$$

$$P(X=4 \text{ and } X=5) = P(X=4) \times P(X=5)$$

$$P(X=4 \text{ or } X=5) = P(X=4) + P(X=5)$$

# Is packet switching always better?

## Great for bursty data

- resource sharing
- simpler, no call setup

## Excessive congestion possible

- packet **delay** and **loss**
- protocols needed for reliable data transfer, congestion control

## How to provide circuit-like behavior?

- bandwidth guarantees needed for audio/video apps
- still an unsolved problem (chapter 7)

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