Chapter 1 Introduction

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James F. Kurose | Keith W. Ross COMPUTER A TOP-DOWN APPROACH P

Computer Networking: A Top-Down Approach

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Chapter 1: roadmap

- What is the Internet?
- What is a protocol?
- Network edge: hosts, access network, physical media
- Network core: packet/circuit switching, internet structure
- Performance: loss, delay, throughput
- Security
- Protocol layers, service models
- History



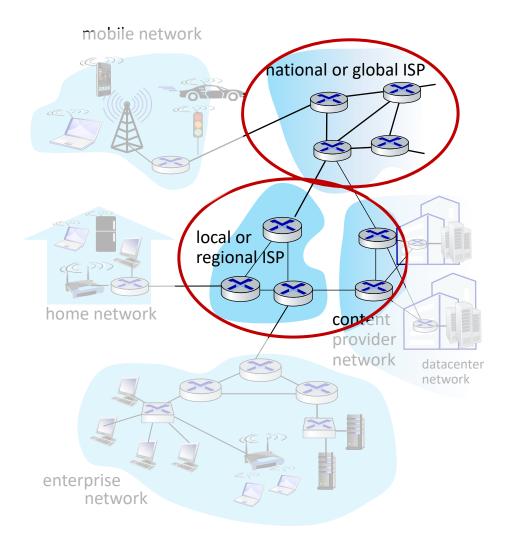
Chapter 1: roadmap

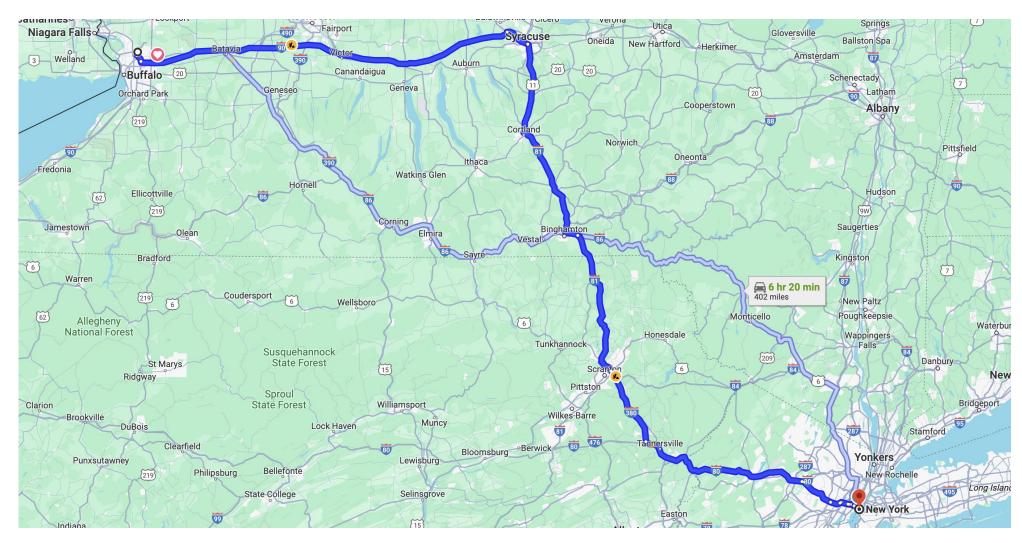
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The network core

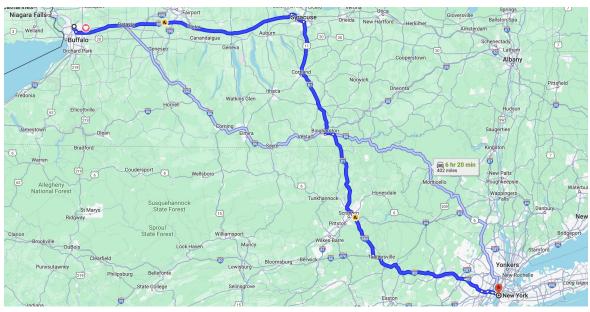
- mesh of interconnected routers
- packet-switching: hosts break application-layer messages into packets
 - network forwards packets from one router to the next, across links on path from source to destination



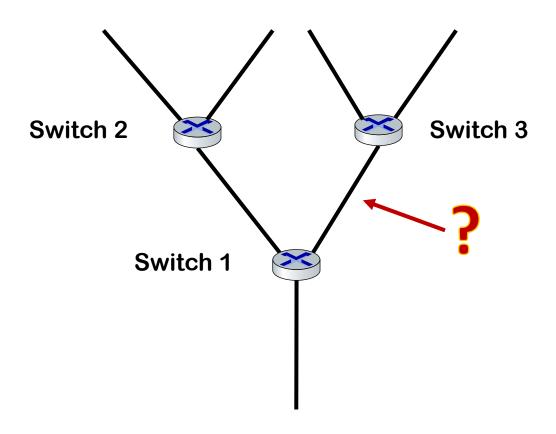






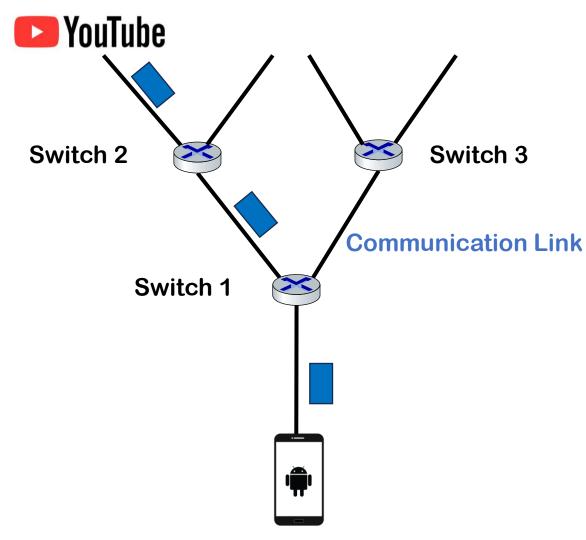


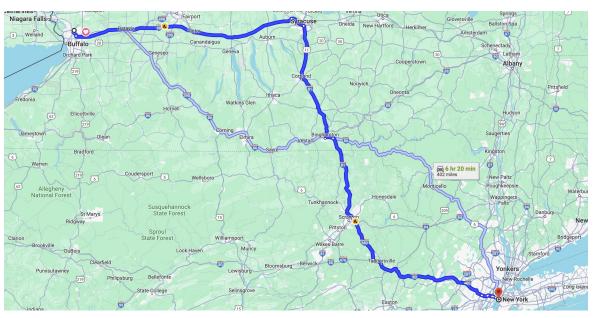


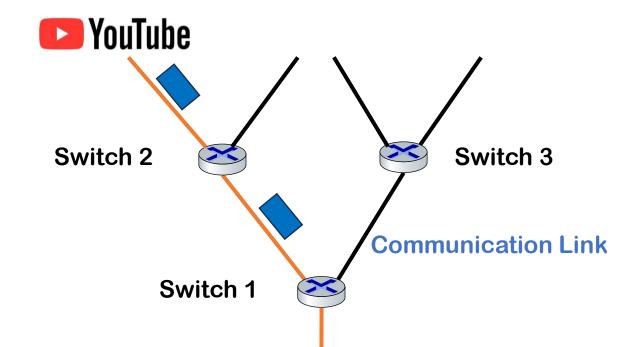












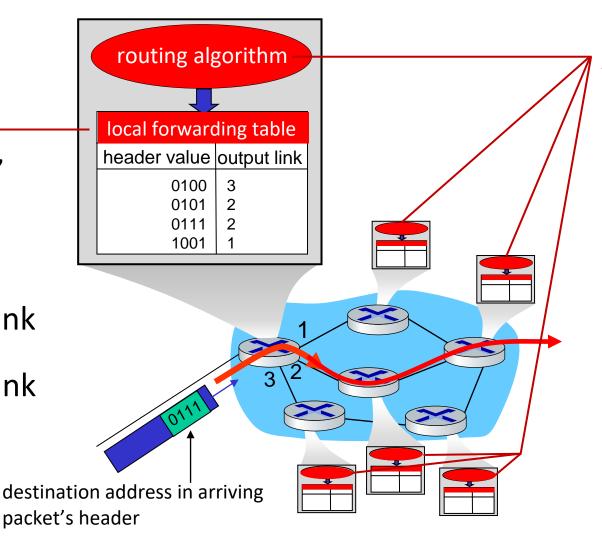


- One packet in and one packet out
- The packet goes to the correct path

Two key network-core functions

Forwarding:

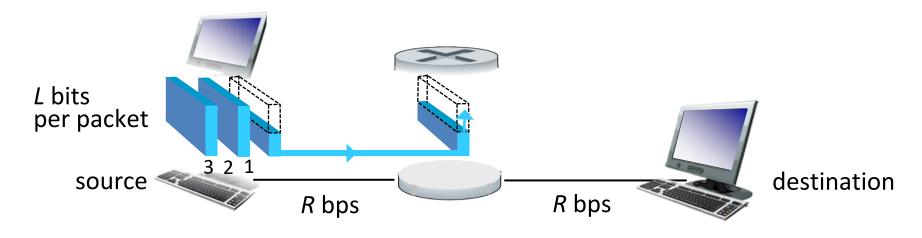
- aka "switching"
- local action: move arriving packets from router's input link to appropriate router output link



Routing:

- global action: determine sourcedestination paths taken by packets
- routing algorithms

Packet-switching: store-and-forward

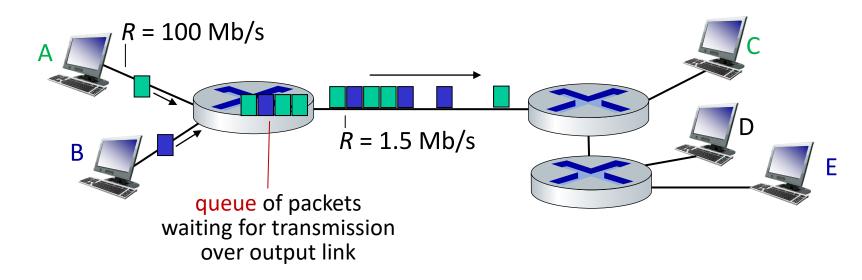


- packet transmission delay: takes L/R seconds to transmit (push out) L-bit packet into link at R bps
- store and forward: entire packet must arrive at router before it can be transmitted on next link

One-hop numerical example:

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

Packet-switching: queueing



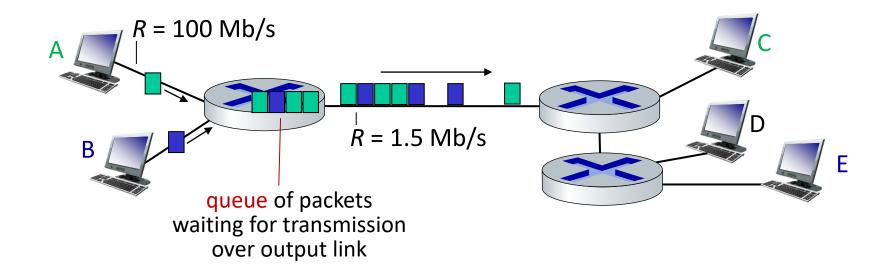
Queueing occurs when work arrives faster than it can be serviced:







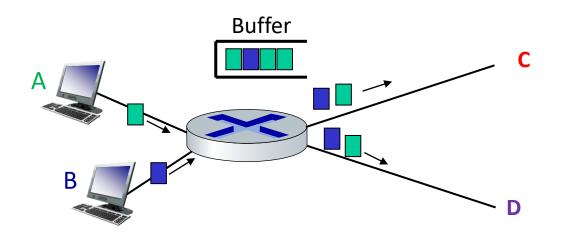
Packet-switching: queueing



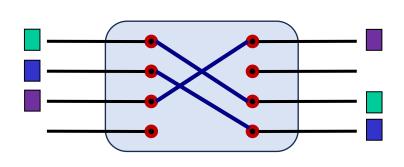
Packet queuing and loss: if arrival rate (in bps) to link exceeds transmission rate (bps) of link for some period of time:

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

Alternative to packet switching: circuit switching (e.g., plain old telephone networks or POTS)





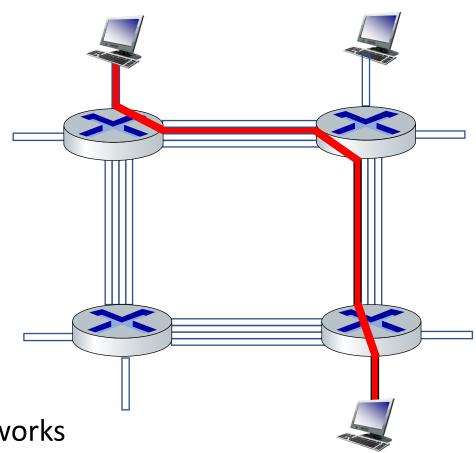


Circuit Switching

Alternative to packet switching: circuit switching (e.g., plain old telephone networks or POTS)

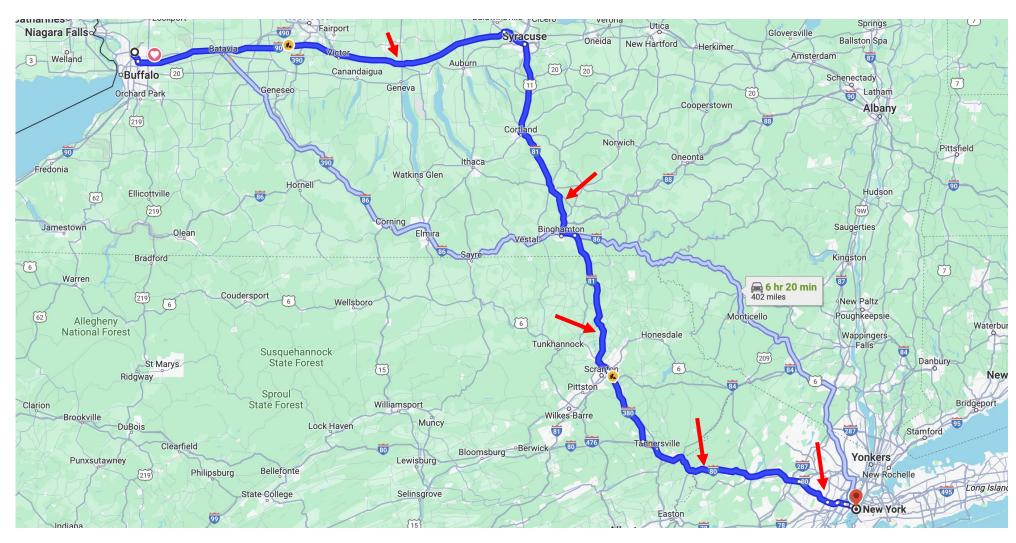
end-end resources allocated to, reserved for "call" between source and destination

- 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



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

Network Switch VS Circuit Switch



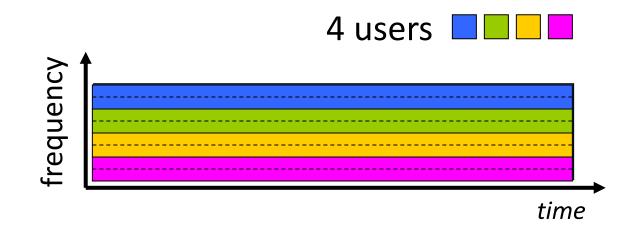
Circuit switching: FDM and TDM

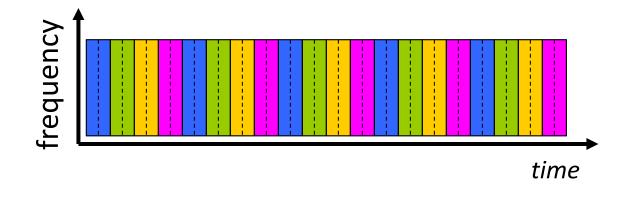
Frequency Division Multiplexing (FDM)

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

Time Division Multiplexing (TDM)

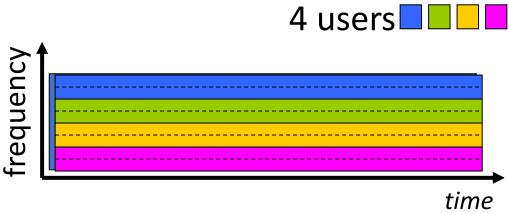
- time divided into slots
- each call allocated periodic slot(s), can transmit at a max rate of (wider) frequency band (only) during its time slot(s)





Road lanes - Multiplexing





Internet Core: Packet Switching

each end-end data stream divided into packets

- users A through C packets share network resources
- each packet uses full link bandwidth
- resources used as needed

Bandwidth division into "pieces"

Dedicated allocation

Resource reservation

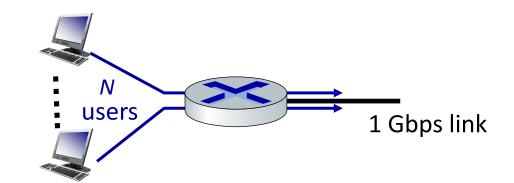
- each packet has a "header" (containing e.g., destination address) in addition to "payload" (data)
- Store and Forward (requires buffer and introduces delay)
- Statistical multiplexing (support an aggregate demand that exceeds link bandwidth)

1-47

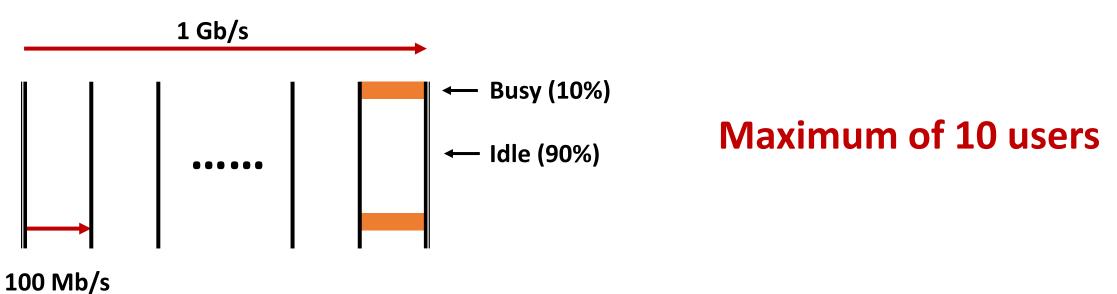
Packet switching versus circuit switching

example:

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



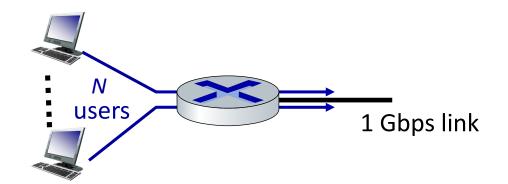
Q: how many users can use this network under circuit-switching and packet switching?



Packet switching versus circuit switching

example:

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



Q: how many users can use this network under circuit-switching and packet switching?

- 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?

A: assuming random arrivals

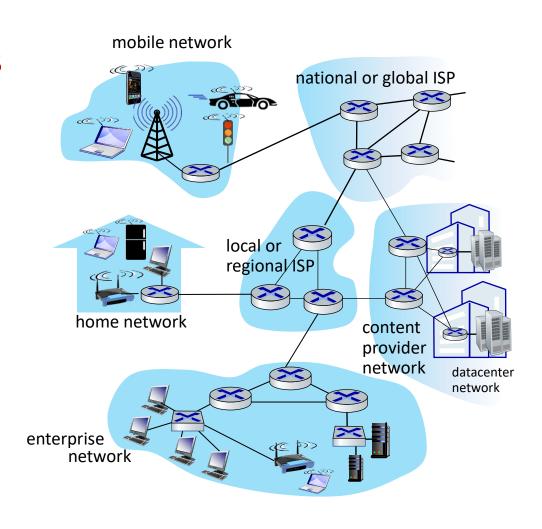
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Packet switching versus circuit switching

Is packet switching a "slam dunk winner"?

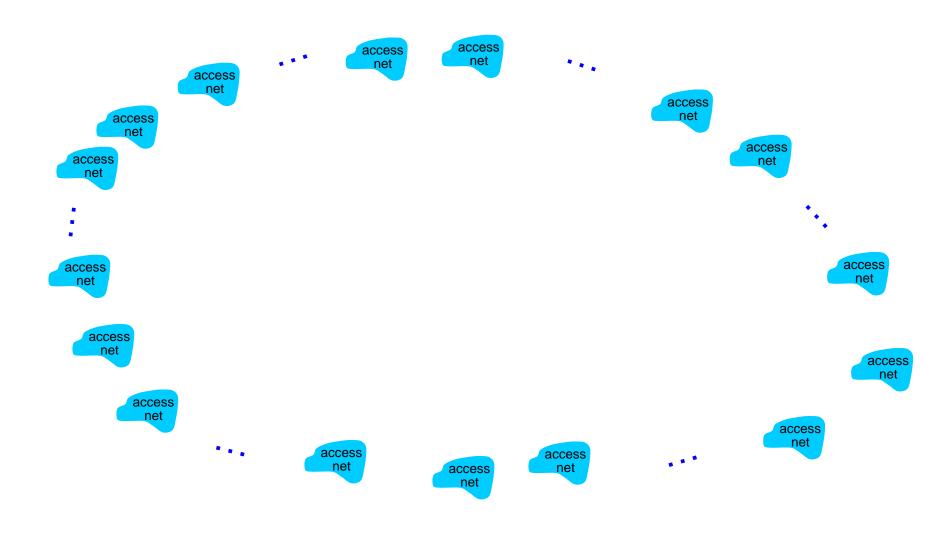
- great for "bursty" data sometimes has data to send, but at other times not
 - resource sharing
 - simpler, no call setup
- excessive congestion possible: packet delay and loss due to buffer overflow
 - protocols needed for reliable data transfer, congestion control
- Q: How to provide circuit-like behavior with packet-switching?
 - "It's complicated." We'll study various techniques that try to make packet switching as "circuit-like" as possible.

- hosts connect to Internet via access Internet Service Providers (ISPs)
- access ISPs in turn must be interconnected
 - so that *any* two hosts (anywhere!) can send packets to each other
- resulting network of networks is very complex
 - evolution driven by economics, national policies

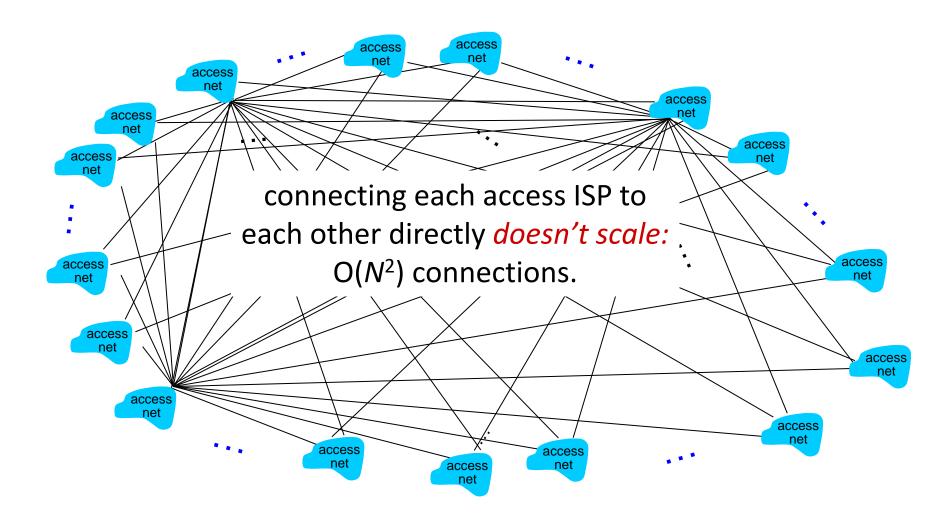


Let's take a stepwise approach to describe current Internet structure

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

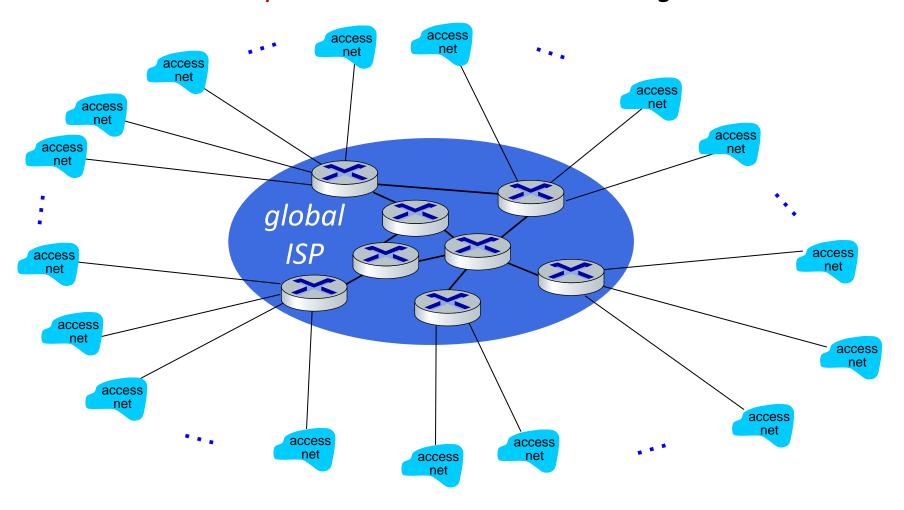


Question: given millions of access ISPs, how to connect them together?

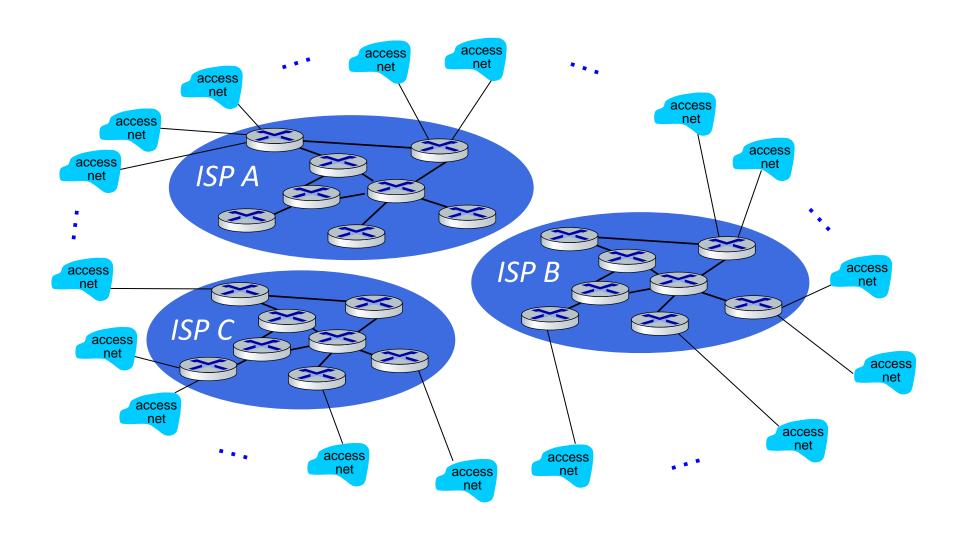


Option: connect each access ISP to one global transit ISP?

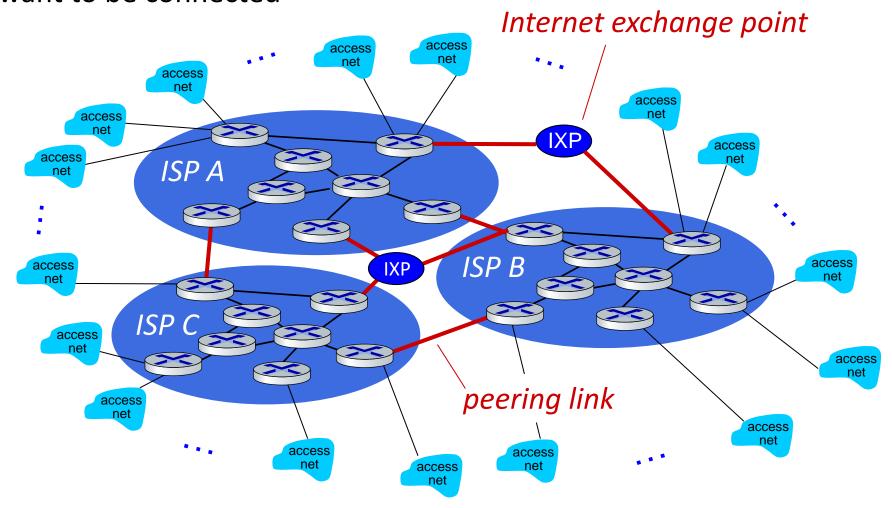
Customer and provider ISPs have economic agreement.



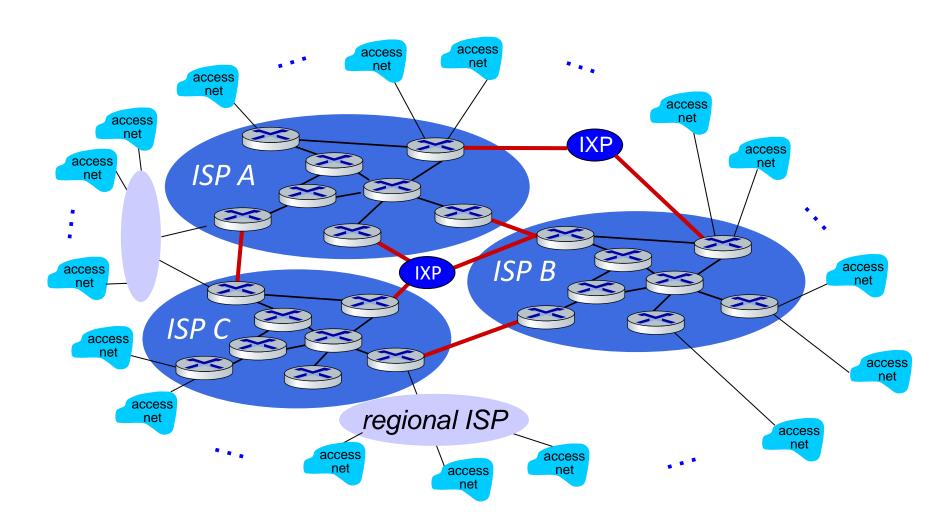
But if one global ISP is viable business, there will be competitors



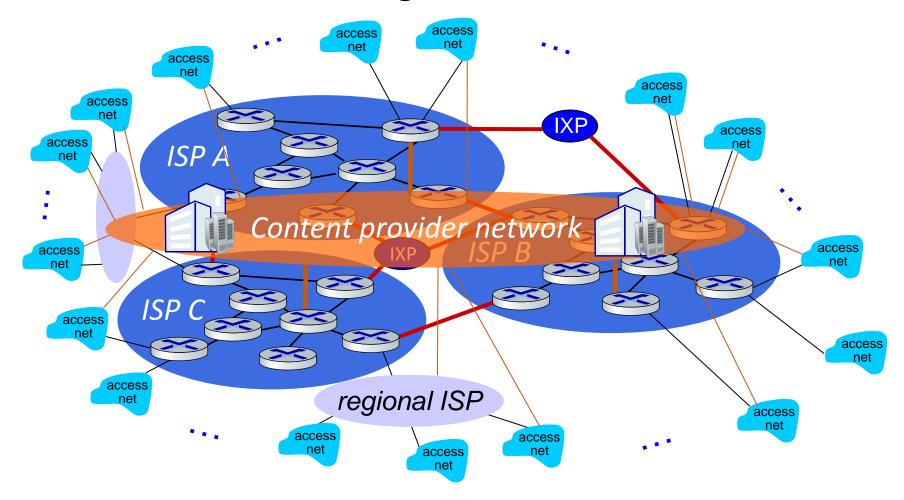
But if one global ISP is viable business, there will be competitors who will want to be connected

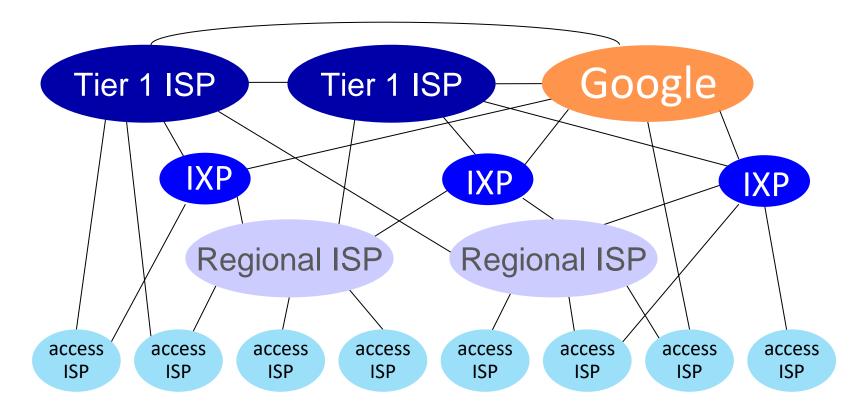


... and regional networks may arise to connect access nets to ISPs



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

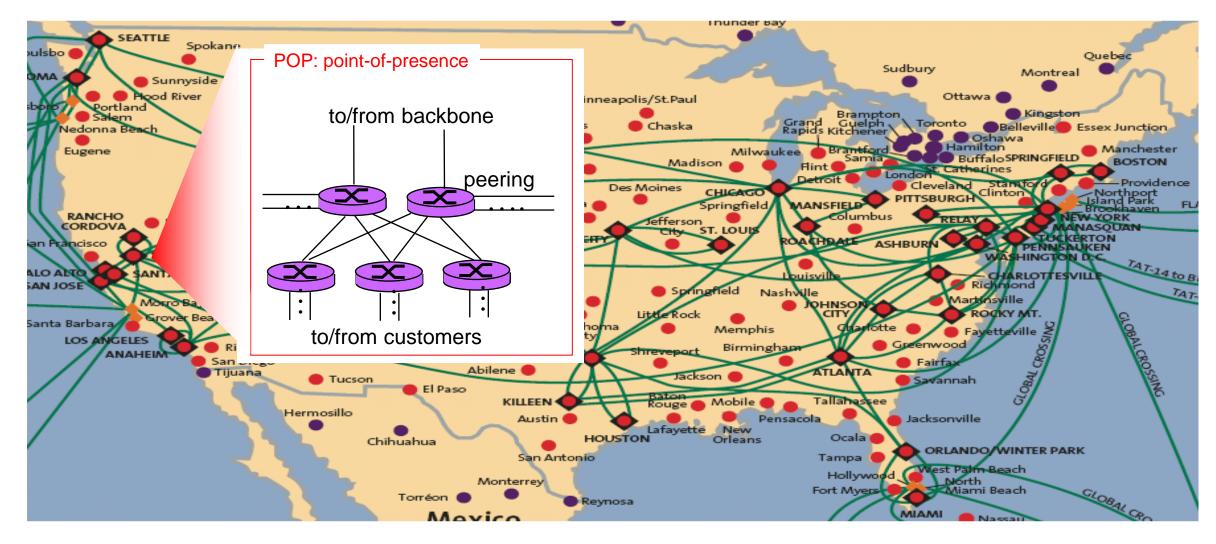




At "center": small # of large but well-connected networks

- "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

Tier-1 ISP: e.g., Sprint/T-Mobile



Introduction

POPs from different Tier-1 ISP connect to each other at IXPs – residing at a building like this in London



Internet Core Routers (including those at POPs/IXP)



Router on "paper"



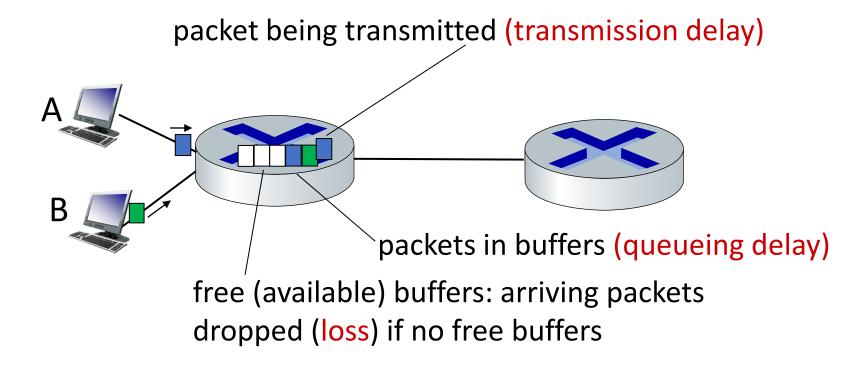
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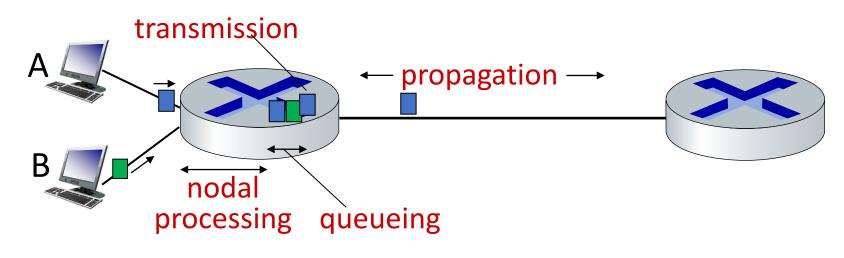


How do packet delay and loss occur?

- packets queue in router buffers, waiting for its turn to be transmitted
 - queue length grows when arrival rate to link (temporarily) exceeds output link capacity
- packet loss occurs when memory to hold queued packets fills up



Packet delay: four sources



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

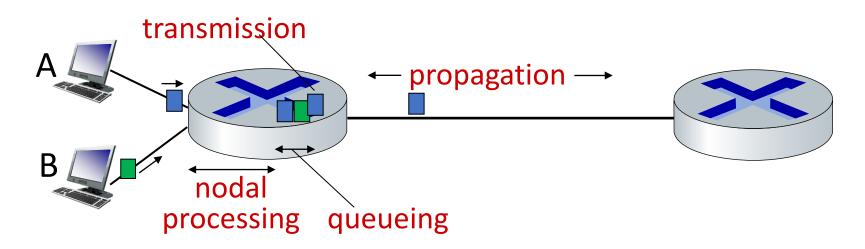
d_{proc} : nodal processing

- check bit errors
- determine output link
- typically < microsecs</p>

d_{queue}: queueing delay

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

Packet delay: four sources



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

d_{trans} : transmission delay:

- L: packet length (bits)
- R: link transmission rate (bps)

$$d_{trans} = L/R$$

$$d_{trans} \text{ and } d_{prop}$$

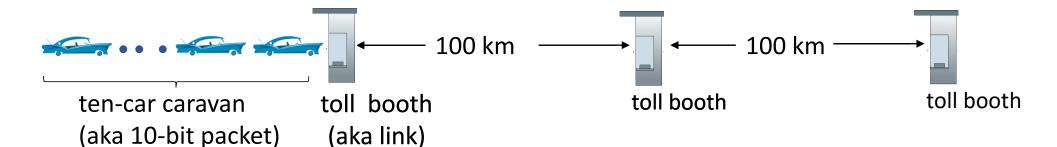
$$very \text{ different}$$

d_{prop} : propagation delay:

- *d*: length of physical link
- s: propagation speed (~2x10⁸ m/sec)

$$d_{\text{prop}} = d/s$$

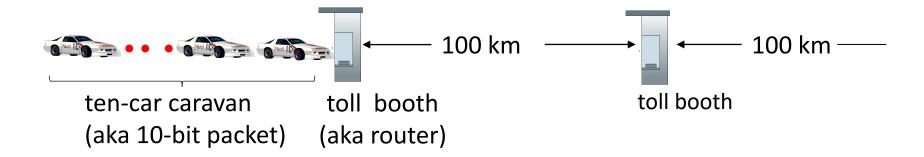
Caravan analogy



- car ~ bit; caravan ~ packet; toll service ~ link transmission
- toll booth takes 12 sec to service car (bit transmission time)
- "propagate" at 100 km/hr
- 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: 100km/(100km/hr) = 1 hr
- A: 62 minutes

Caravan analogy



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

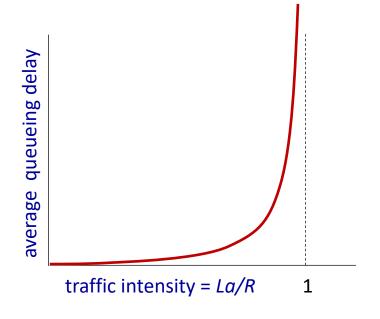
<u>A: Yes!</u> after 7 min, first car arrives at second booth; three cars still at first booth

Packet queueing delay (revisited)

- a: average packet arrival rate
- L: packet length (bits)
- R: link bandwidth (bit transmission rate)

$$\frac{L \cdot a}{R}$$
: arrival rate of bits "traffic service rate of bits intensity"

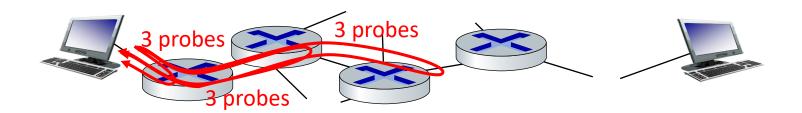
- La/R ~ 0: avg. queueing delay small
- La/R -> 1: avg. queueing delay large
- La/R > 1: more "work" arriving is more than can be serviced - average delay infinite!





"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



Real Internet delays and routes

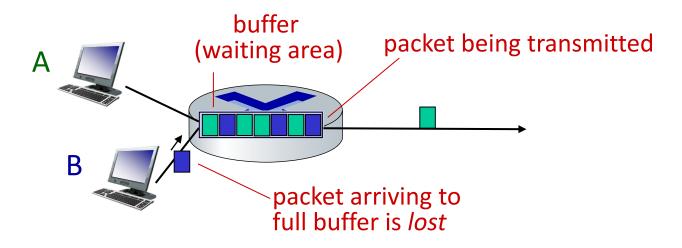
traceroute: gaia.cs.umass.edu to www.eurecom.fr

```
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
1 to border1-rt-fa5-1-0.gw.umass.edu (128.119.3.130) 6 ms 5 ms
                                                                                  to border1-rt-fa5-1-0.gw.umass.edu
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 trans-oceanic link
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
                      * means no response (probe lost, router not replying)
19 fantasia.eurecom.fr (193.55.113.142) 132 ms 128 ms 136 ms
```

^{*} Do some traceroutes from exotic countries at www.traceroute.org

Packet loss

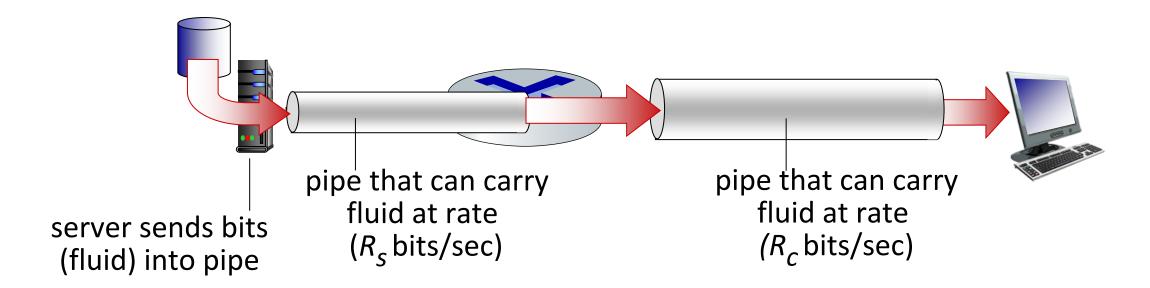
- queue (aka buffer) preceding link in buffer has finite capacity
- packet arriving at a full queue dropped (aka lost)
- lost packet may be retransmitted by previous node, by source end system, or not at all



^{*} Check out the Java applet for an interactive animation (on publisher's website) of queuing and loss

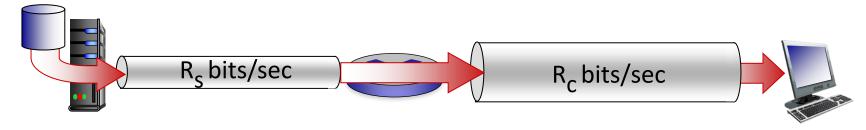
Throughput

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

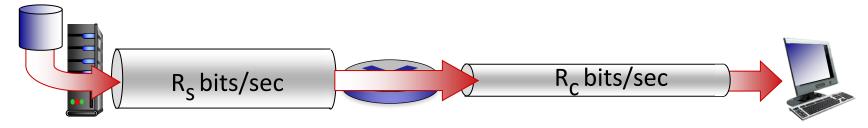


Throughput

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



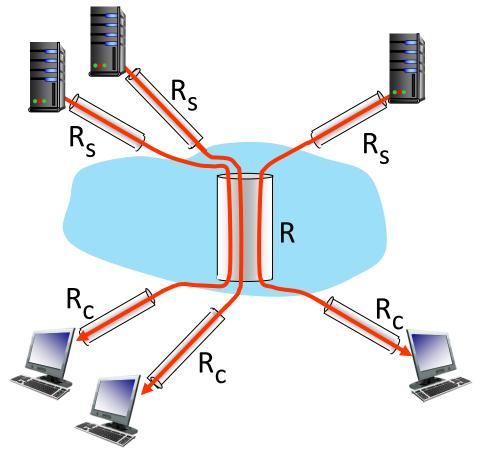
 $R_s > R_c$ What is average end-end throughput?



bottleneck link

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

Throughput: network scenario



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

- per-connection endend 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/