

Computer Networks and the Internet (Cont'd)

Lecture-2

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HOUSEKEEPING & ACKNOWLEDGEMENT



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- Original material can be found on: https://gaia.cs.umass.edu/kurose ross/ppt.htm



Chapter 1: Introduction roadmap

- 1.1 What is the Internet?
- 1.2 Network edge
 - end systems, access networks, links

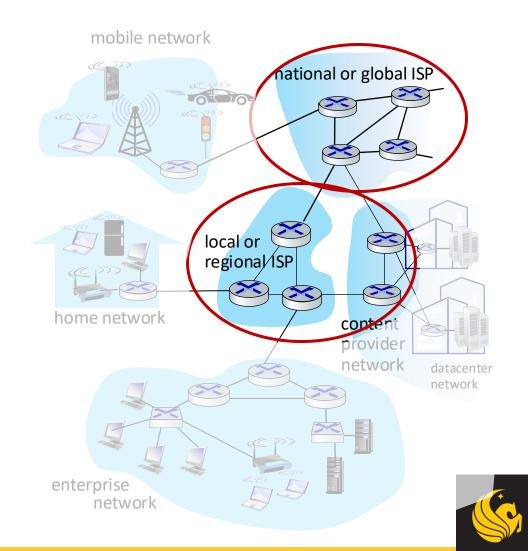
1.3 Network core

- Packet switching, circuit switching, network structure
- 1.4 Delay, loss, throughput in networks
- 1.5 Protocol layers, service models
- 1.6 Networks under attack: security
- 1.7 History



The Network Core

- 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

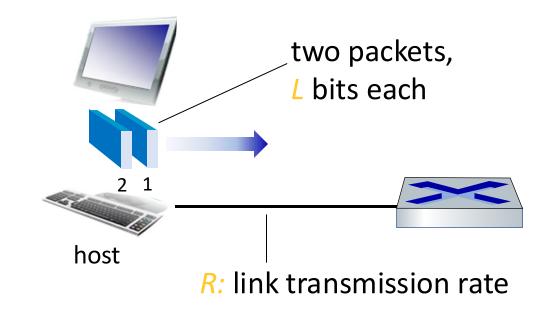


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Host: sends packets of data (reminder)

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





packet time needed to transmission = transmit *L*-bit delay packet into link

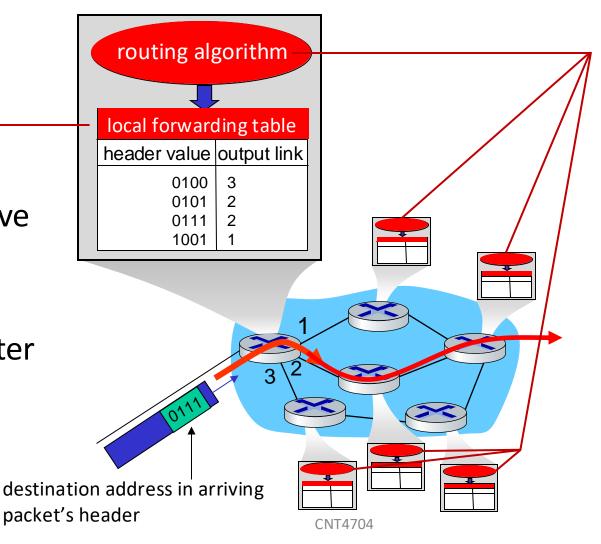
$$= \frac{L}{R}^{\text{(bits)}}$$
(bits/sec)



Two key network-core functions

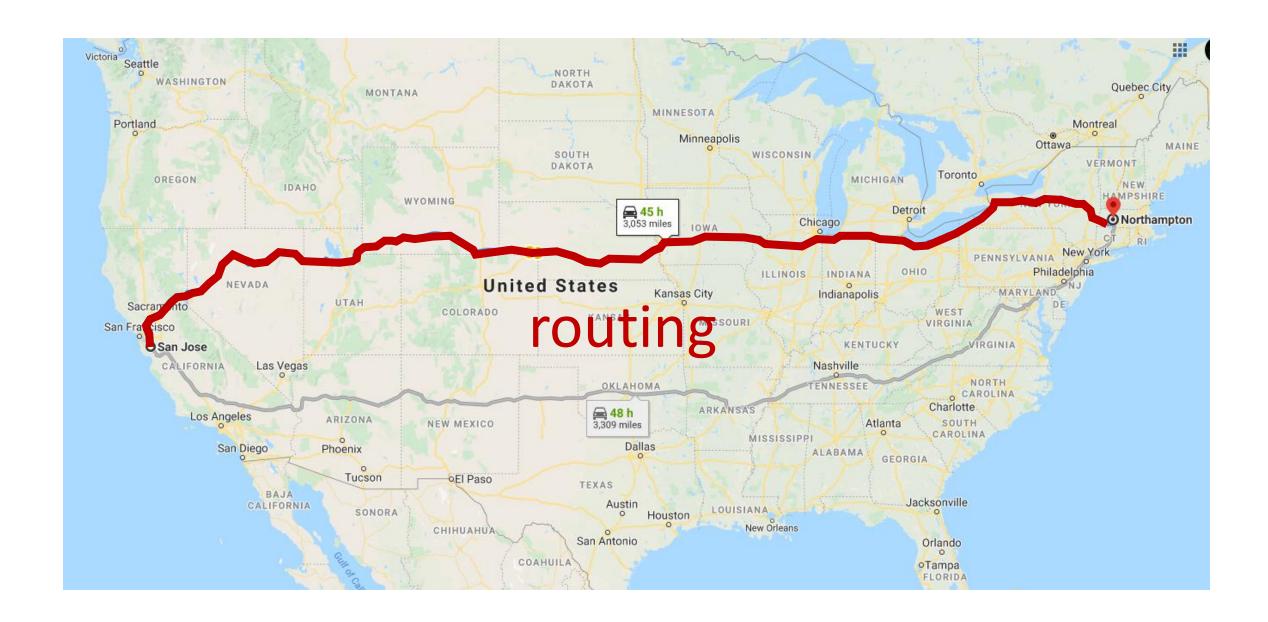
Forwarding:

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



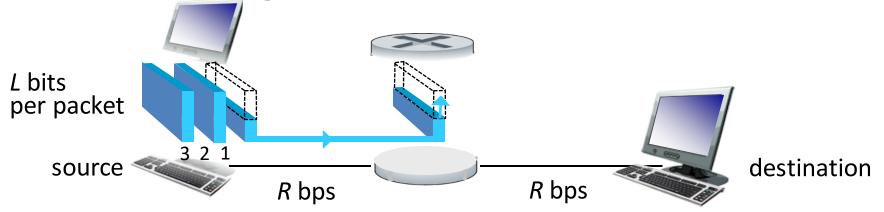
Routing:

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





Packet-switching: store-and-forward



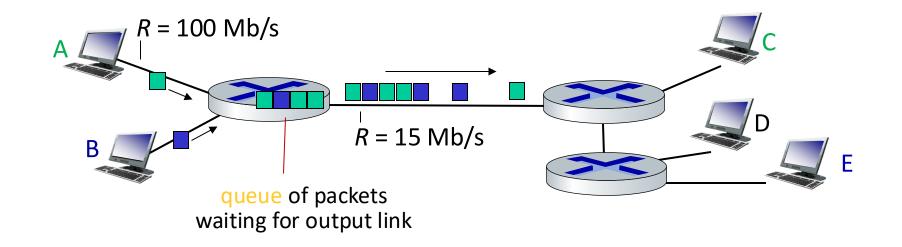
- Transmission delay: takes L/R seconds to transmit (push out) L-bit packet into link at R bps
- Store and forward: entire packet must arrive at router before it can be transmitted on next link

One-hop numerical example:

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



Packet-switching: queueing delay, loss



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

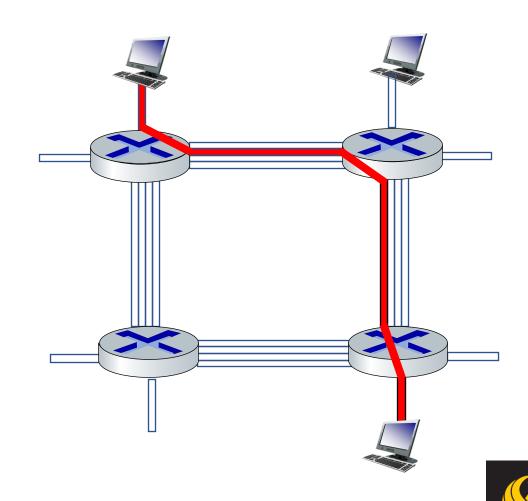
- packets will queue, waiting to be transmitted on output link
- packets can be dropped (lost) if memory (buffer) in router fills up



Alternative to packet switching: circuit switching

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

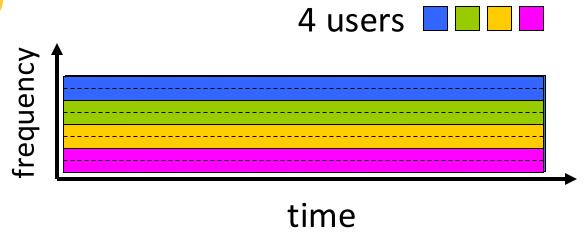
- in diagram, each link has four circuits.
 - call gets 2nd circuit in top link and 1st circuit in right link.
- dedicated resources: no sharing
 - circuit-like (guaranteed) performance
- circuit segment idle if not used by call (no sharing)
- commonly used in traditional telephone networks



Circuit Switching: FDM and TDM

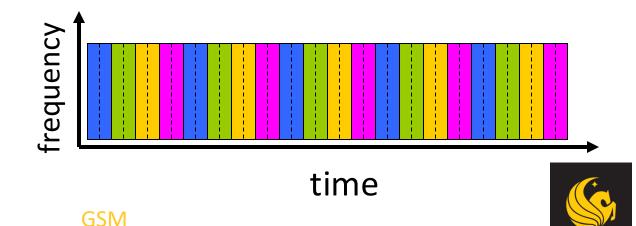
Frequency Division Multiplexing (FDM)

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

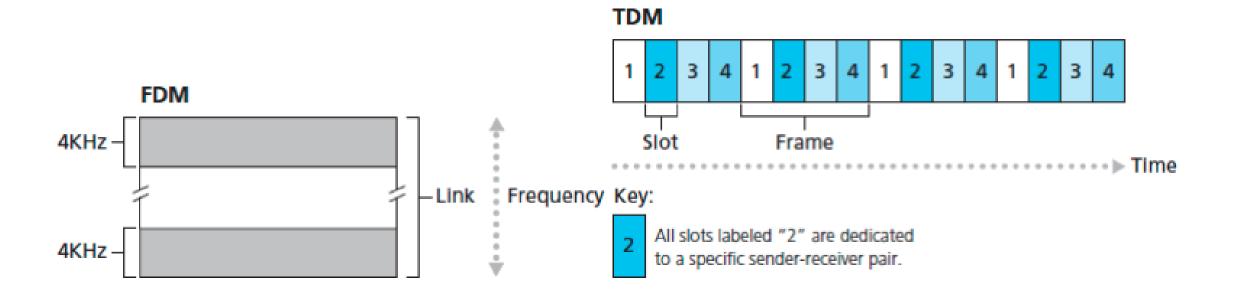


Time Division Multiplexing (TDM)

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



Circuit Switching: FDM and TDM



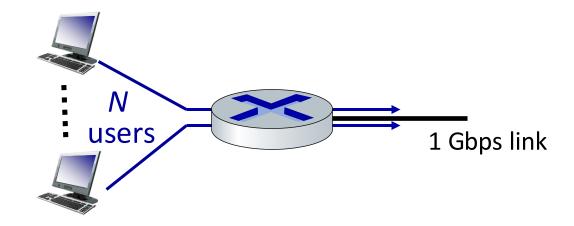


Packet switching versus circuit switching

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
- packet switching: with 35 users, probability > 10 active at same time is less than .0004 *



Q: how did we get value 0.0004?

A: Hw problem



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

Packet switching versus circuit switching

Is packet switching a "slam dunk winner"?

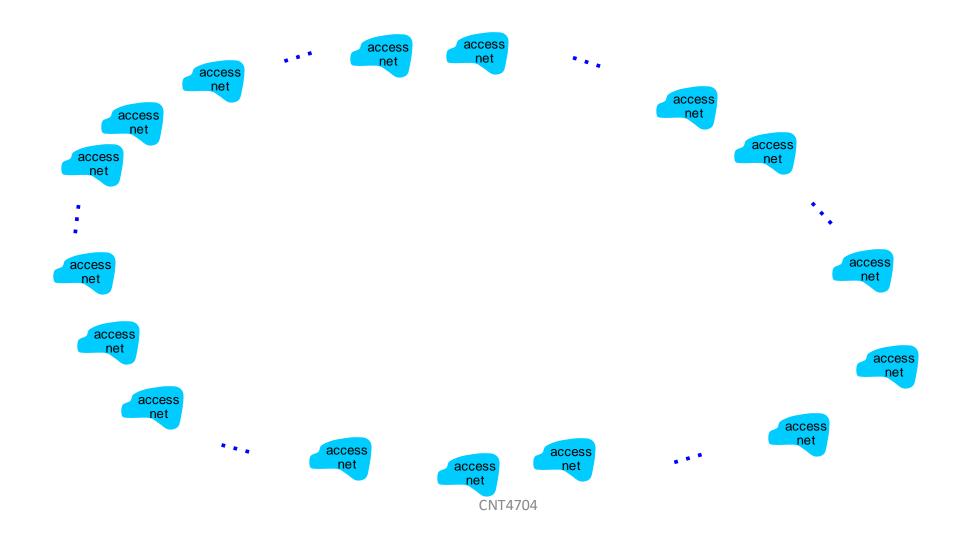
- great for "bursty" data sometimes has data to send, but at other times not
 - resource sharing
 - simpler, no call setup
- excessive congestion possible: packet delay and loss due to buffer overflow
 - protocols needed for reliable data transfer, congestion control
- Q: How to provide circuit-like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



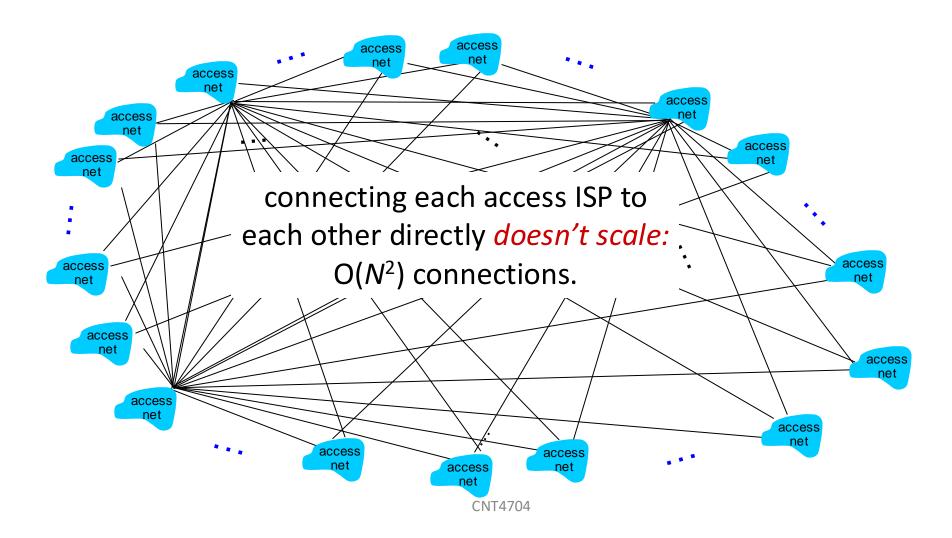
- Hosts connect to Internet via access Internet Service Providers (ISPs)
 - residential, enterprise (company, university, commercial) ISPs
- Access ISPs in turn must be interconnected
 - so that any two hosts can send packets to each other
- Resulting network of networks is very complex
 - evolution was driven by economics and national policies
- Let's take a stepwise approach to describe current Internet structure



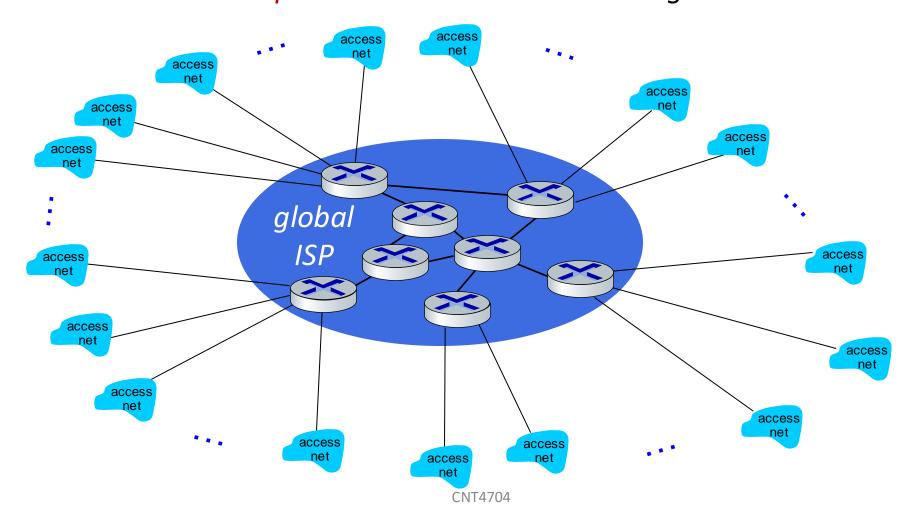
Question: given millions of access ISPs, how to connect them together?



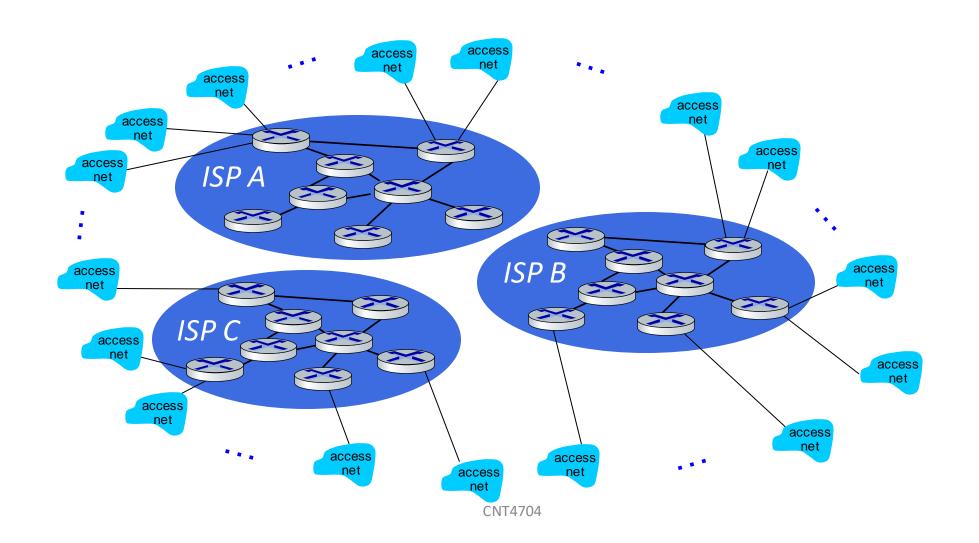
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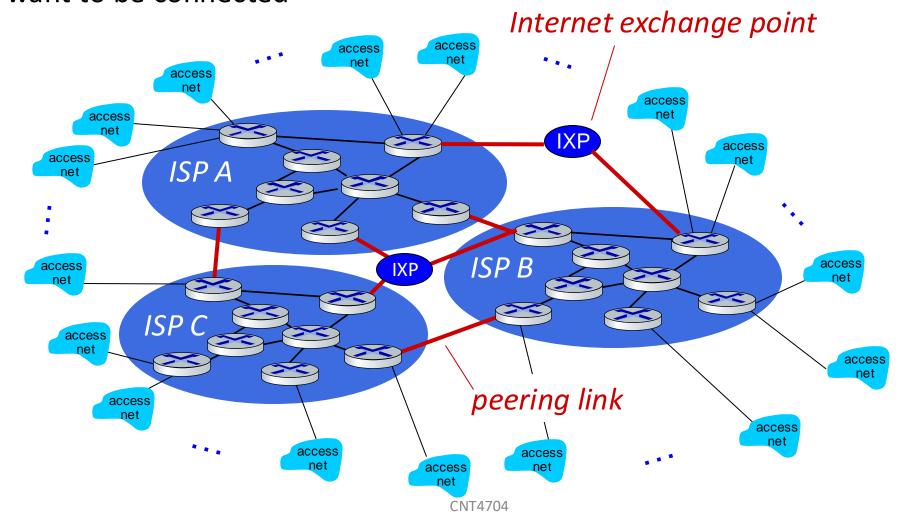
Option: connect each access ISP to one global transit ISP? Customer and provider ISPs have economic agreement.



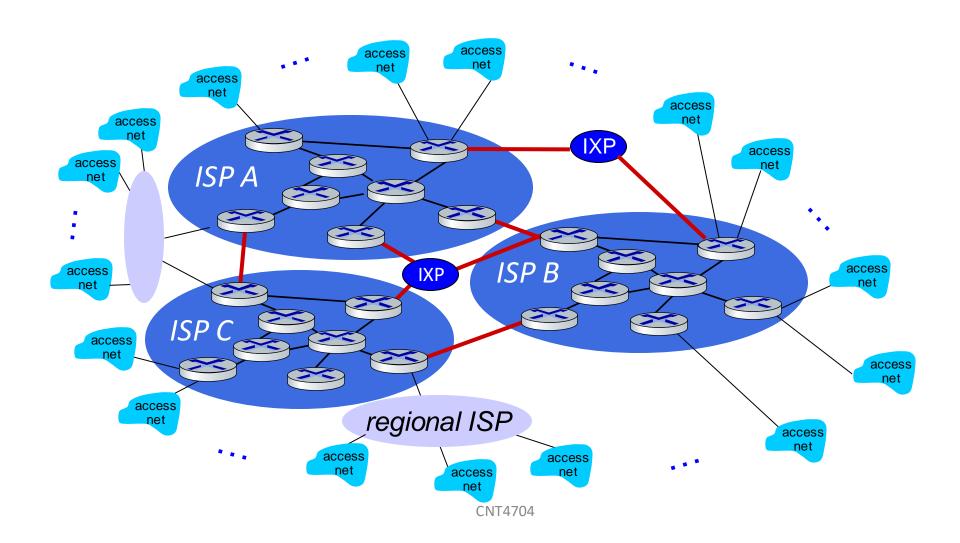
But if one global ISP is viable business, there will be competitors



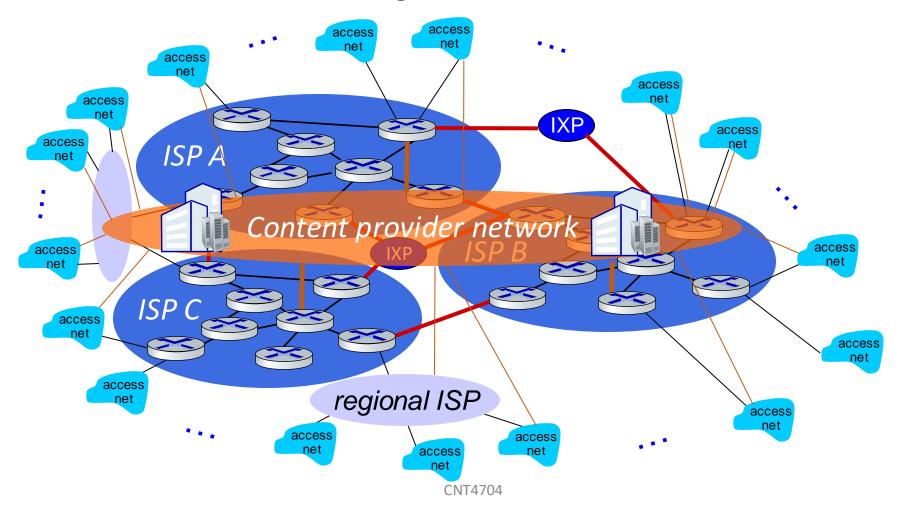
But if one global ISP is viable business, there will be competitors who will want to be connected

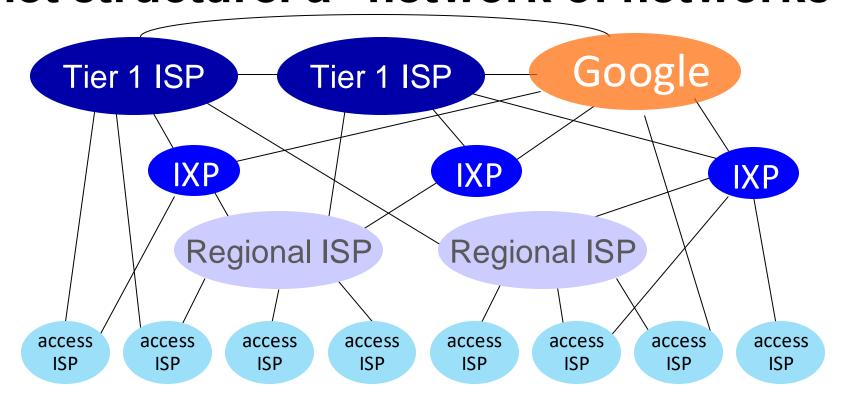


... and regional networks may arise to connect access nets to ISPs



... 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 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



CHAPTER 1: INTRODUCTION ROADMAP

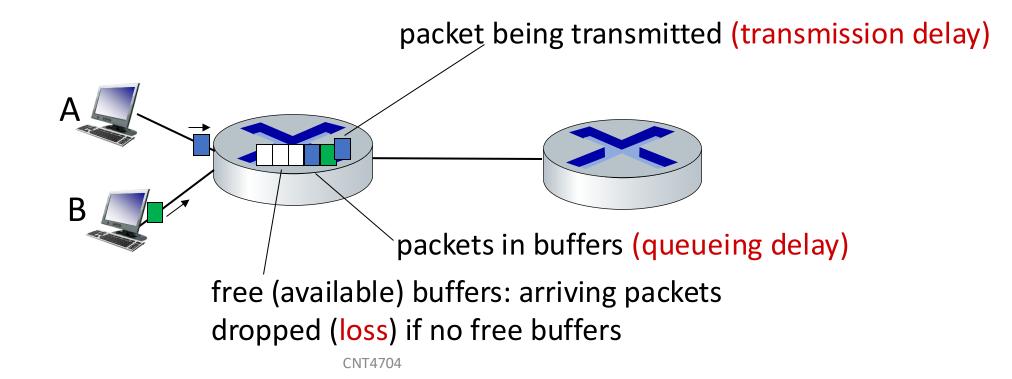
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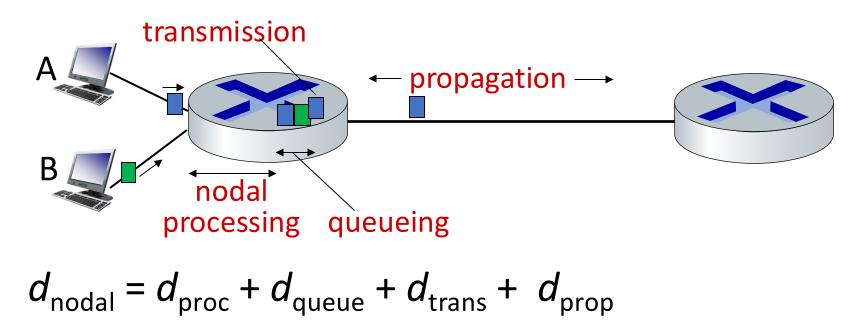
How do packet loss and delay occur?

packets queue in router buffers

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



Packet delay: four sources



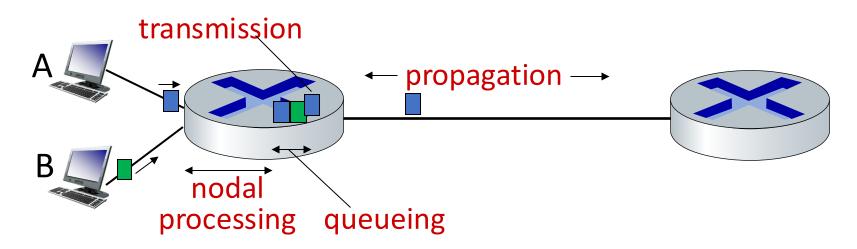
d_{proc} : nodal processing

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

$$\frac{d_{trans} = L/R}{d_{trans}}$$
 and $\frac{d_{prop}}{very}$ different

d_{prop} : propagation delay:

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

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

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

Questions?



