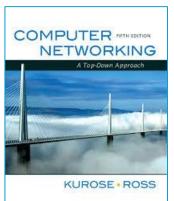
CSC358 Intro. to Computer Networks

Lecture 5: Review, Transport Layer, (de)multiplexing, UDP, reliable data transfer

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Office Hours: T 17:00–18:00 R 9:00–10:00 BA4222

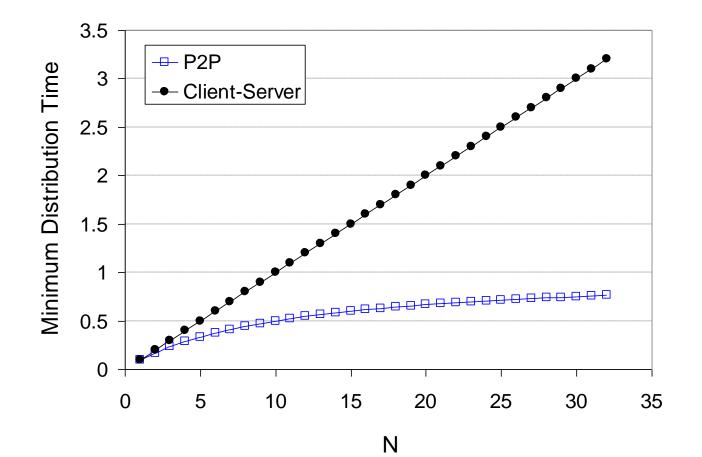
TA Office Hours: W 16:00-17:00 BA3201 R 10:00-11:00 BA7172 csc358ta@cdf.toronto.edu http://www.cs.toronto.edu/~ahchinaei/teaching/2016jan/csc358/

Review

- Examples of Application Layer Protocols:
 - TLS, DNS, μTP
 - Proprietary protocols: e.g. Skype or myGame
- Examples of Transport Layer Protocols:
 - TCP, UDP, DCCP
- Examples of Network Layer Protocols:
 - IP, ICMP
- DNS (TCP/UDP)P2P

Client-server vs. P2P: example

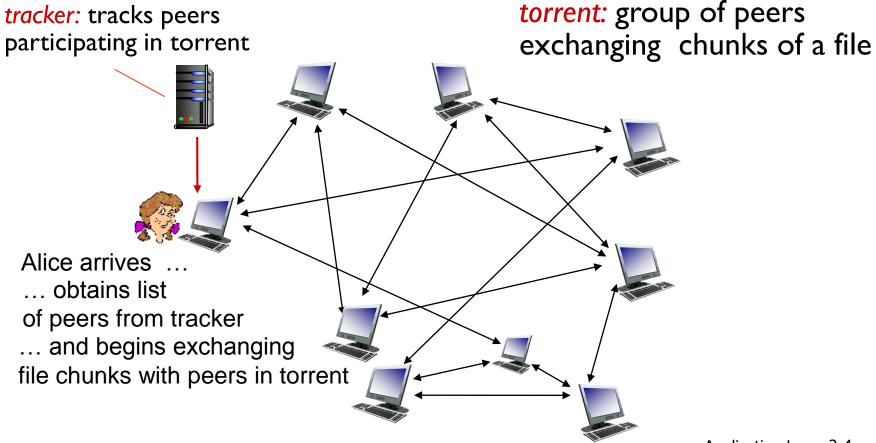
client upload rate = u, F/u = 1 hour, $u_s = 10u$, $d_{min} \ge u_s$



Application Layer 2-3

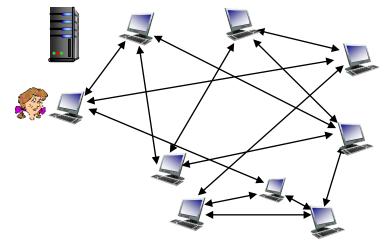
P2P file distribution: BitTorrent

- file divided into 256Kb chunks
- peers in torrent send/receive file chunks



P2P file distribution: BitTorrent

- peer joining torrent:
 - has no chunks, but will accumulate them over time from other peers
 - registers with tracker to get list of peers, connects to subset of peers ("neighbors")

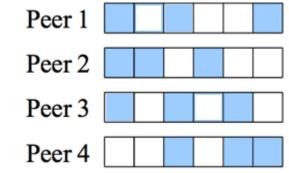


- while downloading, peer uploads chunks to other peers
- peer may change peers with whom it exchanges chunks
- churn: peers may come and go
- once peer has entire file, it may (selfishly) leave or (altruistically) remain in torrent

BitTorrent: requesting, sending file chunks

requesting chunks:

- at any given time, different peers have different subsets of file chunks
- periodically, Alice asks each peer for list of chunks that they have
- Alice requests missing chunks from peers, rarest first



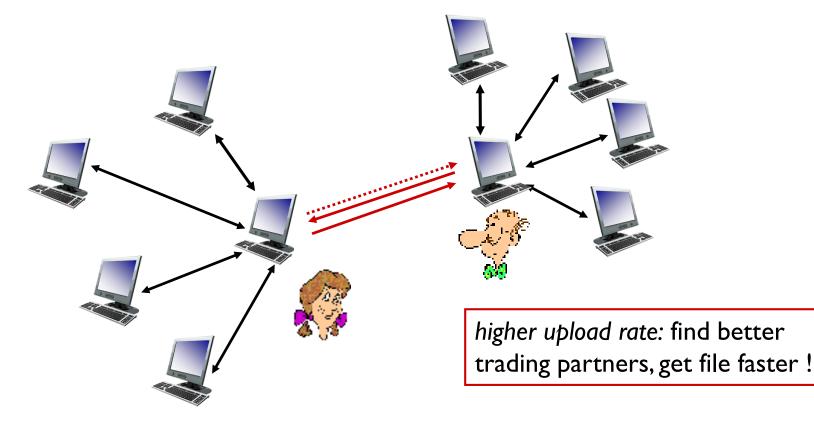
sending chunks: tit-for-tat

- Alice sends chunks to those four peers currently sending her chunks at highest rate
 - other peers are choked by Alice (do not receive chunks from her)
 - re-evaluate top 4 every 10 secs
- every 30 secs: randomly select another peer, starts sending chunks
 - "optimistically unchoke" this peer
 - newly chosen peer may join top 4

BitTorrent: tit-for-tat

(I) Alice "optimistically unchokes" Bob

- (2) Alice becomes one of Bob's top-four providers; Bob reciprocates
- (3) Bob becomes one of Alice's top-four providers



Distributed Hash Table (DHT)

- Hash table
- DHT paradigm
- Circular DHT and overlay networks
- Peer churn

Simple Database

Simple database with(key, value) pairs:

key: human name; value: social security #

Key	Value
John Washington	132-54-3570
Diana Louise Jones	761-55-3791
Xiaoming Liu	385-41-0902
Rakesh Gopal	441-89-1956
Linda Cohen	217-66-5609
Lisa Kobayashi	177-23-0199

• key: movie title; value: IP address

Hash Table

- More convenient to store and search on numerical representation of key
- key = hash(original key)

Original Key	Key	Value
John Washington	8962458	132-54-3570
Diana Louise Jones	7800356	761-55-3791
Xiaoming Liu	1567109	385-41-0902
Rakesh Gopal	2360012	441-89-1956
Linda Cohen	5430938	217-66-5609
Lisa Kobayashi	9290124	177-23-0199

Distributed Hash Table (DHT)

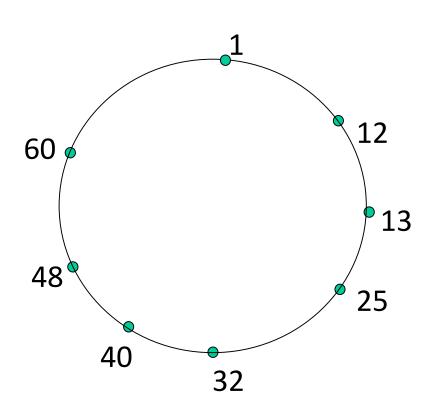
- Distribute (key, value) pairs over millions of peers
 - pairs are evenly distributed over peers
- Any peer can query database with a key
 - database returns value for the key
 - To resolve query, small number of messages exchanged among peers
- Each peer only knows about a small number of other peers
- Robust to peers coming and going (churn)

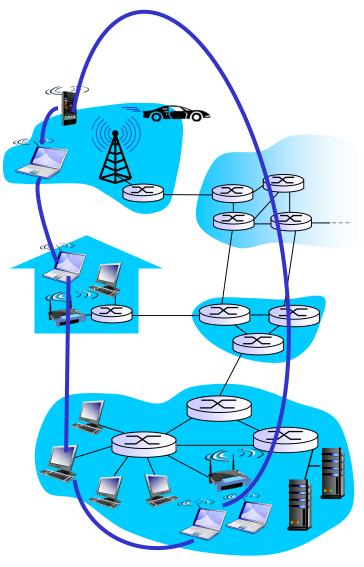
Assign key-value pairs to peers

- rule: assign key-value pair to the peer that has the closest ID.
- convention: closest is the *immediate successor* of the key.
- * e.g., ID space {0,1,2,3,...,63}
- suppose 8 peers: 1,12,13,25,32,40,48,60
 - If key = 51, then assigned to peer 60
 - If key = 60, then assigned to peer 60
 - If key = 61, then assigned to peer 1

Circular DHT

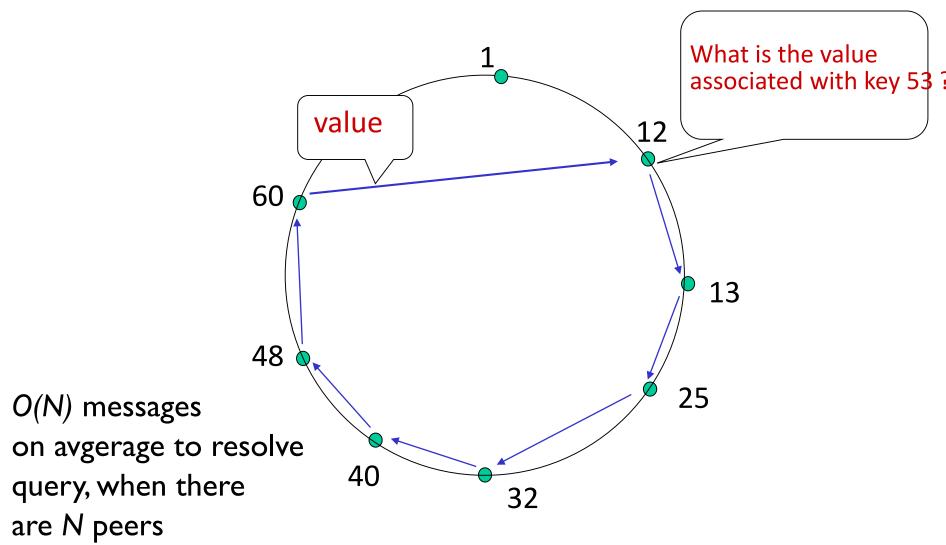
 each peer only aware of immediate successor and predecessor.



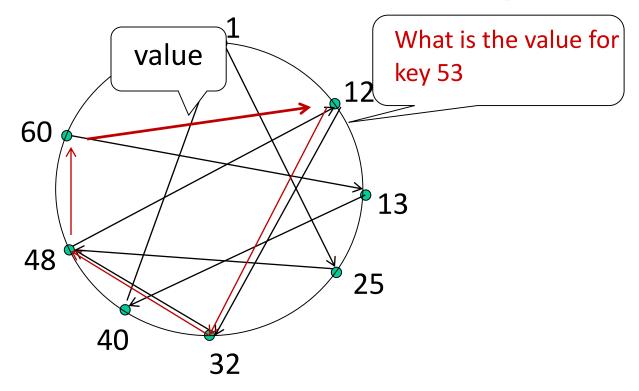


"overlay network"

Resolving a query

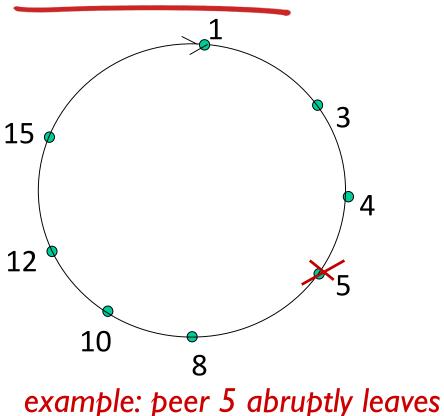


Circular DHT with shortcuts



- each peer keeps track of IP addresses of predecessor, successor, short cuts.
- reduced from 6 to 3 messages.
- possible to design shortcuts with O(log N) neighbors, O(log N) messages in query

Peer churn



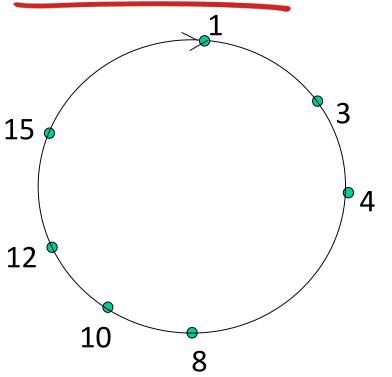
handling peer churn:

peers may come and go (churn)
each peer knows address of its two successors

*each peer periodically pings its
two successors to check aliveness

*if immediate successor leaves, choose next successor as new immediate successor

Peer churn



handling peer churn:

peers may come and go (churn)
each peer knows address of its two successors

*each peer periodically pings its
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*if immediate successor leaves, choose next successor as new immediate successor

example: peer 5 abruptly leaves

*peer 4 detects peer 5' s departure; makes 8 its immediate
successor

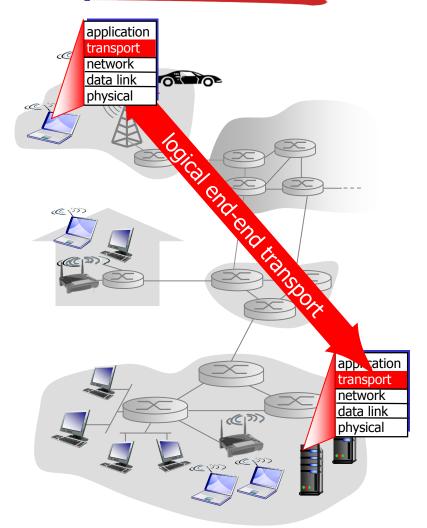
* 4 asks 8 who its immediate successor is; makes 8' s immediate successor its second successor.

Let's move on to Transport Layer

- TCP, UDP
- principles, services
- multiplexing, demultiplexing
- reliable data transfer
- flow control
- congestion control

Transport services and protocols

- provide logical communication between app processes running on different hosts
- transport protocols run in hosts
 - send side: breaks app messages into segments, passes to network layer
 - rcv side: reassembles segments into messages, passes to app layer
- more than one transport protocol available to apps
 - E.g.: TCP, UDP



Transport vs. network layer

- transport layer: logical communication between A. processes
 - relies on, enhances, network layer services
- network layer: logical communication between T. processes
 - .relies on lower layer

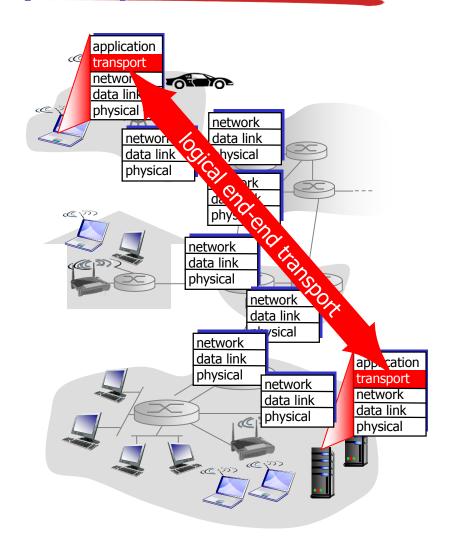
household analogy:

12 kids in Ann's house sending letters to 12 kids in Bill's house:

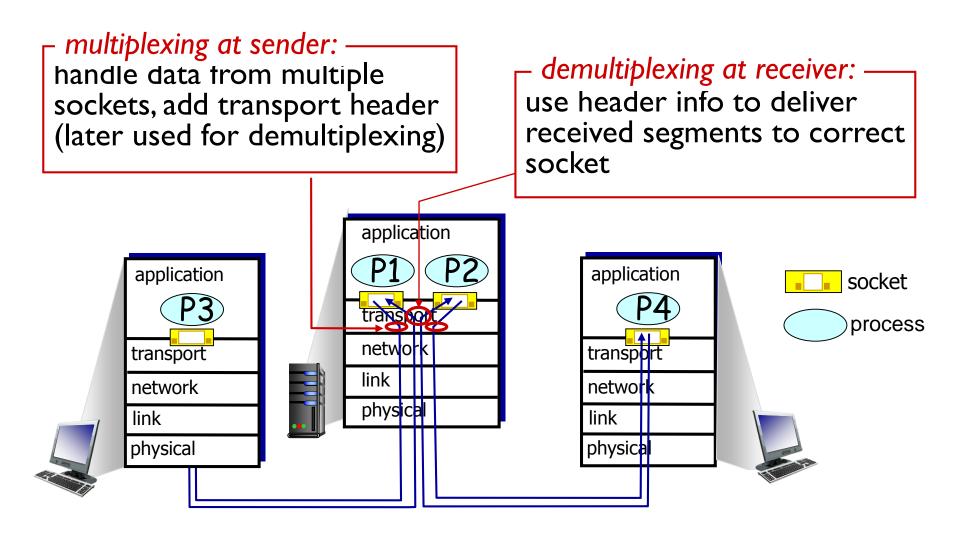
- hosts = houses
- processes = kids
- app messages =
 (continuous) letter
- segments= letter (piece) in envelopes
- transport protocol = Ann and Bill who demux to inhouse siblings
- network-layer protocol = postal service

Internet transport-layer protocols

- reliable, in-order delivery (TCP)
 - congestion control
 - flow control
 - connection setup
- unreliable, unordered delivery: UDP
 - no-frills extension of "best-effort" IP
- services not available:
 - delay guarantees
 - bandwidth guarantees

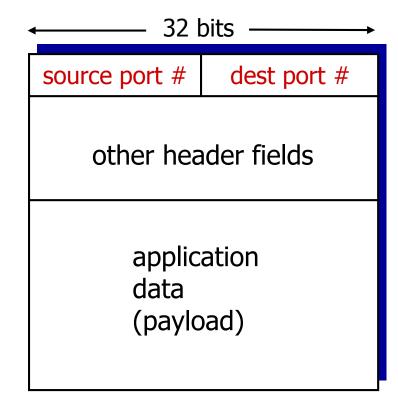


Multiplexing/demultiplexing



How demultiplexing works

- host receives IP datagrams
 - each datagram has source IP address, destination IP address
 - each datagram carries one transport-layer segment
 - each segment has source, destination port number
- host uses IP addresses & port numbers to direct segment to appropriate socket



TCP/UDP segment format

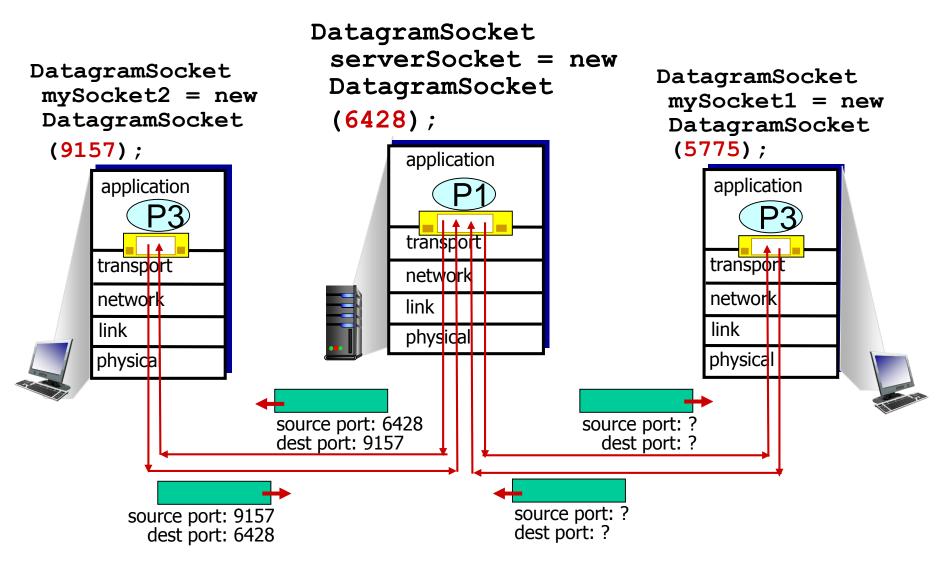
Connectionless demultiplexing

- * recall: created socket has host-local port #: DatagramSocket mySocket1
 - = new DatagramSocket(12534);
- recall: when creating datagram to send into UDP socket, must specify
 - destination IP address
 - destination port #

- when host receives UDP segment:
 - checks destination port # in segment
 - directs UDP segment to socket with that port #

IP datagrams with same dest. port #, but different source IP addresses and/or source port numbers will be directed to same socket at dest

Connectionless demux: example

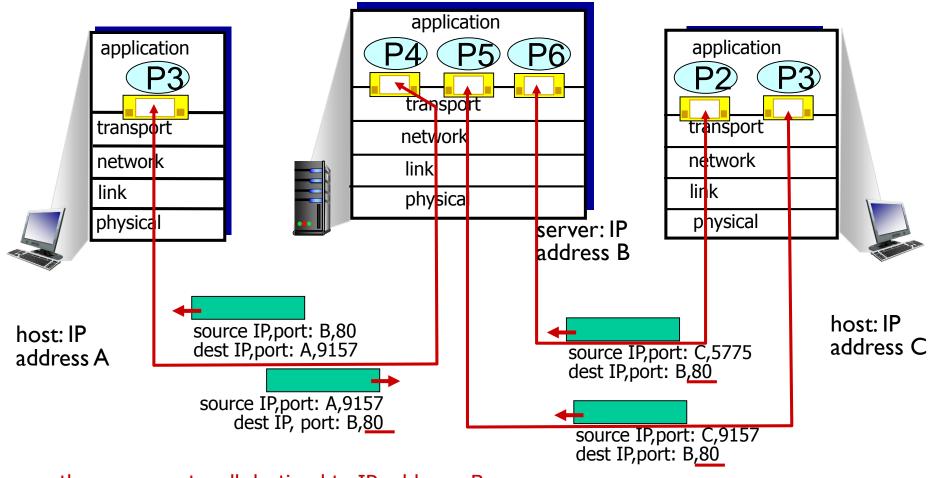


Connection-oriented demux

- TCP socket identified by 4-tuple:
 - source IP address
 - source port number
 - dest IP address
 - dest port number
- demux: receiver uses all four values to direct segment to appropriate socket

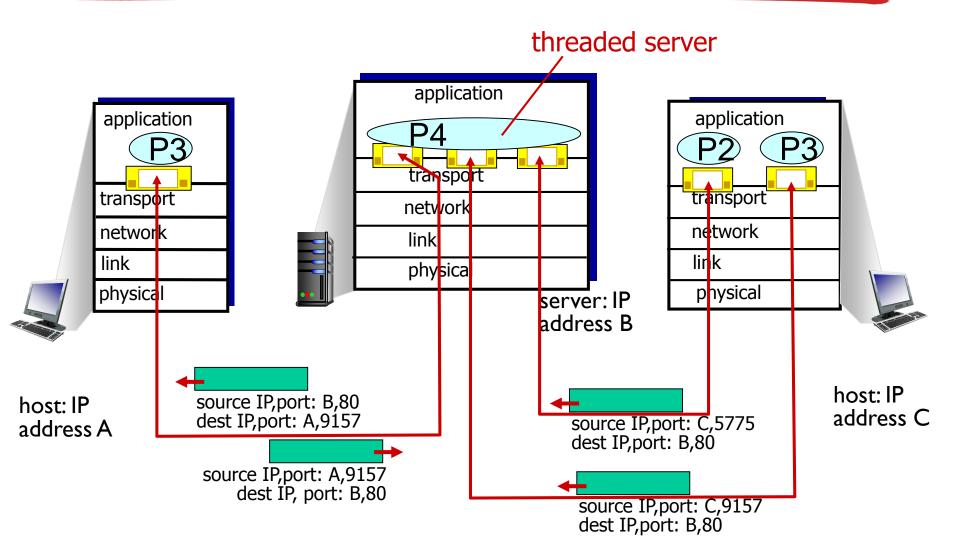
- server host may support many simultaneous TCP sockets:
 - each socket identified by its own 4-tuple
- web servers have different sockets for each connecting client
 - non-persistent HTTP will have different socket for each request

Connection-oriented demux: example



three segments, all destined to IP address: B, dest port: 80 are demultiplexed to *different* sockets

Connection-oriented demux: example

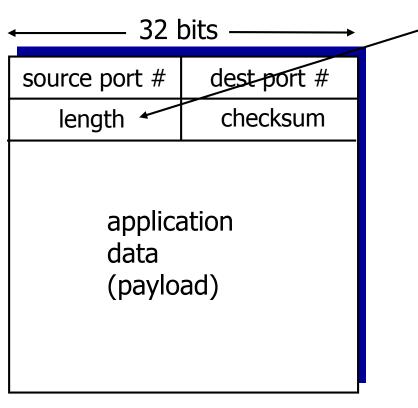


UDP: User Datagram Protocol [RFC 768]

- "no frills," "bare bones" Internet transport protocol
- "best effort" service,
 UDP segments may be:
 - Iost
 - delivered out-of-order to app
- connectionless:
 - no handshaking between UDP sender, receiver
 - each UDP segment handled independently of others

- UDP use:
 - streaming multimedia apps (loss tolerant, rate sensitive)
 - DNS
 - SNMP
- reliable transfer over UDP:
 - add reliability at application layer
 - application-specific error recovery!

UDP: segment header



UDP segment format

length, in bytes of UDP segment, including header

why is there a UDP?

- no connection establishment (which can add delay)
- simple: no connection
 state at sender, receiver
- small header size
- no congestion control: UDP can blast away as fast as desired

UDP checksum

Goal: detect "errors" (e.g., flipped bits) in transmitted segment

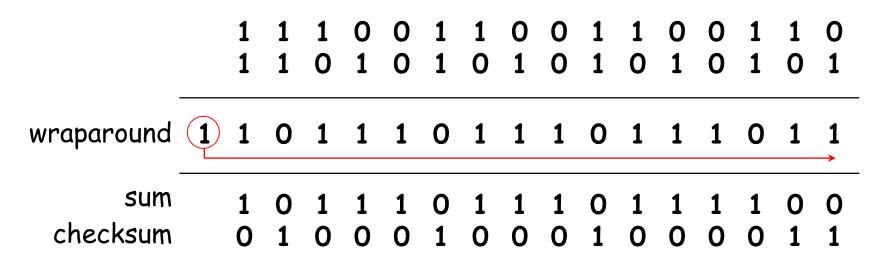
sender:

- treat segment contents, including header fields, as sequence of 16-bit integers
- checksum: addition (one's complement sum) of segment contents
- sender puts checksum value into UDP checksum field

receiver:

- compute checksum of received segment
- check if computed checksum equals checksum field value:
 - NO error detected
 - YES no error detected. But maybe errors nonetheless? More later in Sec 5.2

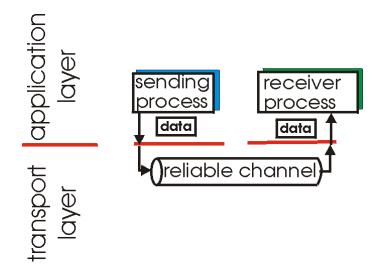
example: add two 16-bit integers



Note: when adding numbers, a carryout from the most significant bit needs to be added to the result

Principles of reliable data transfer

- important in application, transport, link layers
 - top-10 list of important networking topics!

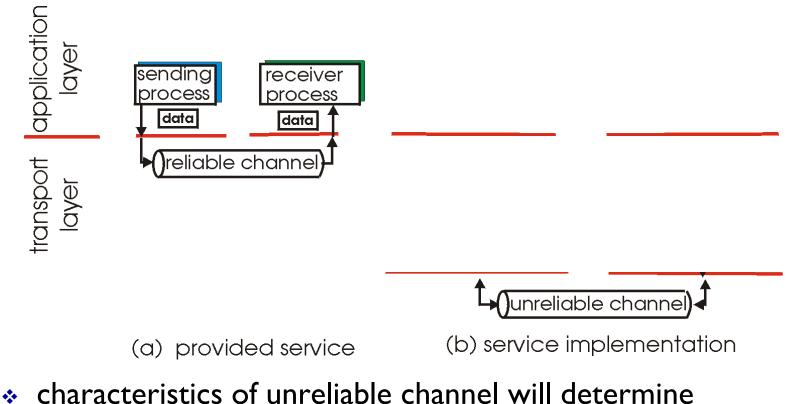


(a) provided service

 characteristics of unreliable channel will determine complexity of reliable data transfer protocol (rdt)

Principles of reliable data transfer

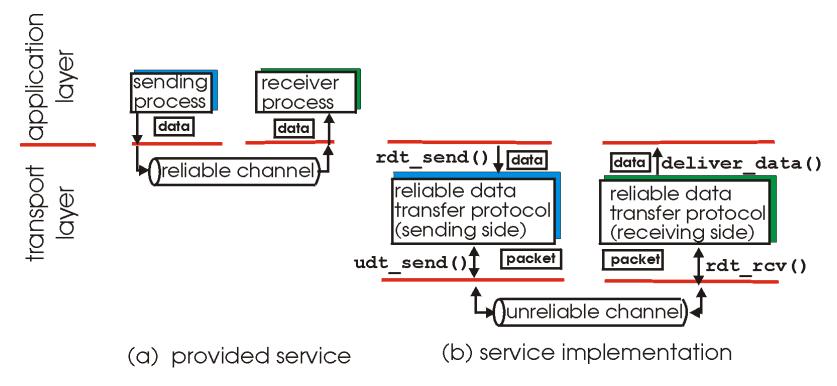
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 characteristics of unreliable channel will determine complexity of reliable data transfer protocol (rdt)

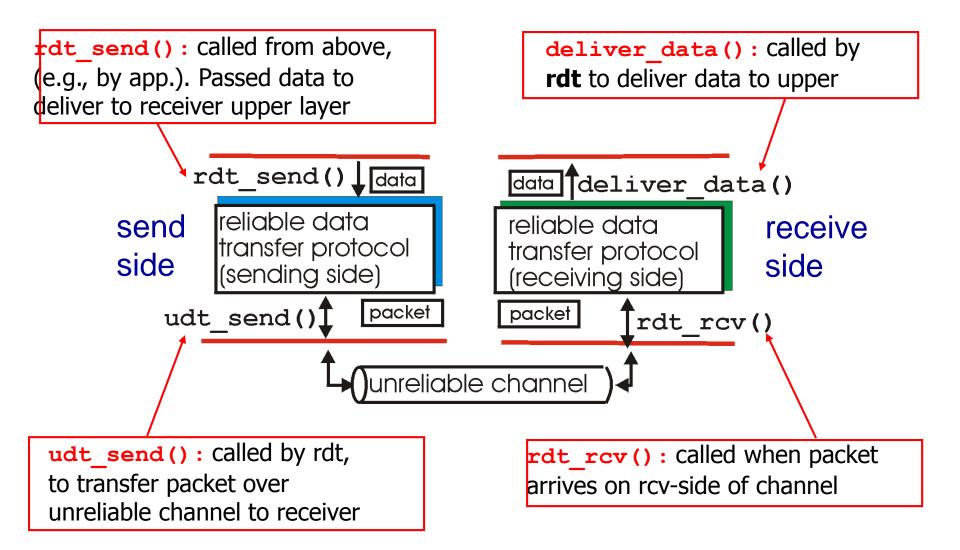
Principles of reliable data transfer

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 characteristics of unreliable channel will determine complexity of reliable data transfer protocol (rdt)

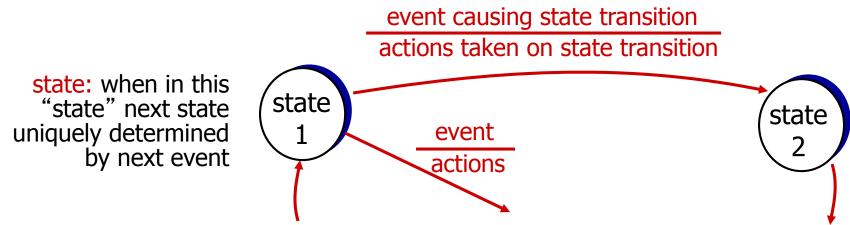
Reliable data transfer: getting started



Reliable data transfer: getting started

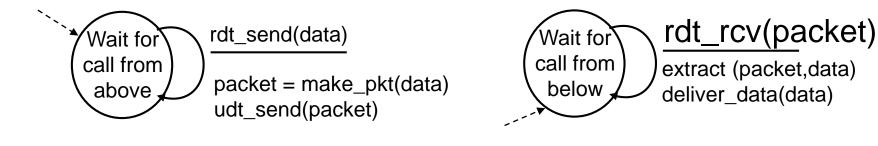
we'll:

- incrementally develop sender, receiver sides of <u>r</u>eliable <u>d</u>ata <u>t</u>ransfer protocol (rdt)
- consider only unidirectional data transfer
 - but control info will flow on both directions!
- use finite state machines (FSM) to specify sender, receiver



rdt1.0: reliable transfer over a reliable channel

- underlying channel perfectly reliable
 - no bit errors
 - no loss of packets
- separate FSMs for sender, receiver:
 - sender sends data into underlying channel
 - receiver reads data from underlying channel



sender

receiver

rdt2.0: over a channel with bit errors

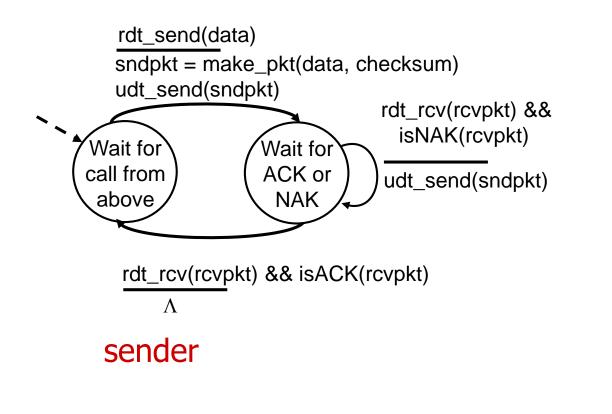
- underlying channel may flip bits in packet
 - checksum to detect bit errors
- * the question: how to recover from errors:

How do humans recover from "errors" during conversation?

rdt2.0: channel with bit errors

- underlying channel may flip bits in packet
 - checksum to detect bit errors
- * the question: how to recover from errors:
 - acknowledgements (ACKs): receiver explicitly tells sender that pkt received OK
 - negative acknowledgements (NAKs): receiver explicitly tells sender that pkt had errors
 - sender retransmits pkt on receipt of NAK
- new mechanisms in rdt2.0 (beyond rdt1.0):
 - error detection
 - feedback: control msgs (ACK,NAK) from receiver to sender

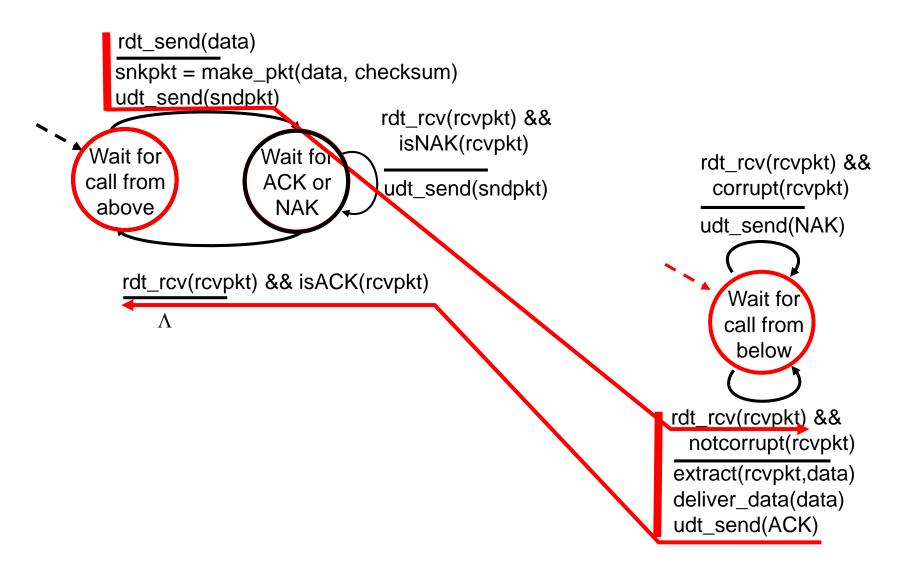
rdt2.0: FSM specification



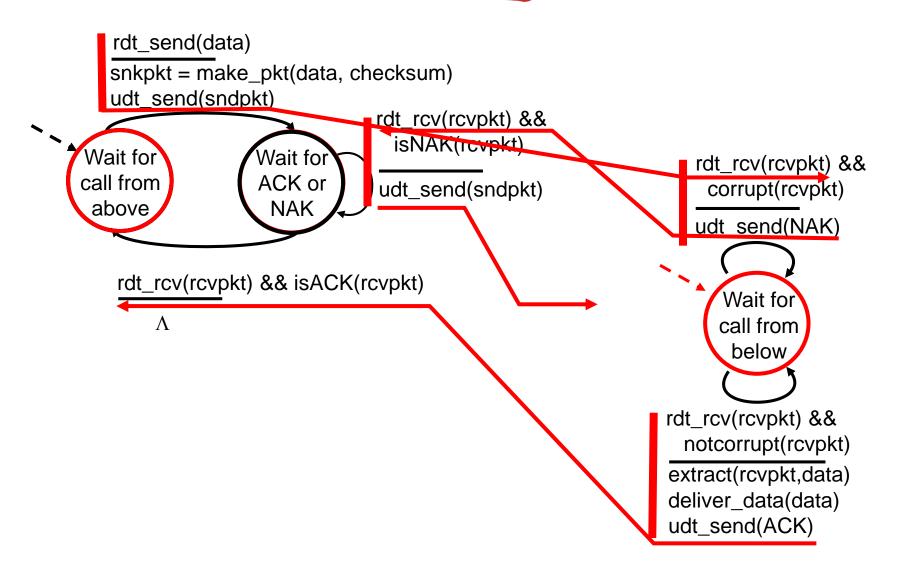
receiver

rdt_rcv(rcvpkt) && corrupt(rcvpkt) udt send(NAK) Wait for call from below rdt_rcv(rcvpkt) && notcorrupt(rcvpkt) extract(rcvpkt,data) deliver_data(data) udt_send(ACK)

rdt2.0: operation with no errors



rdt2.0: error scenario



rdt2.0 has a fatal flaw!

what happens if ACK/NAK corrupted?

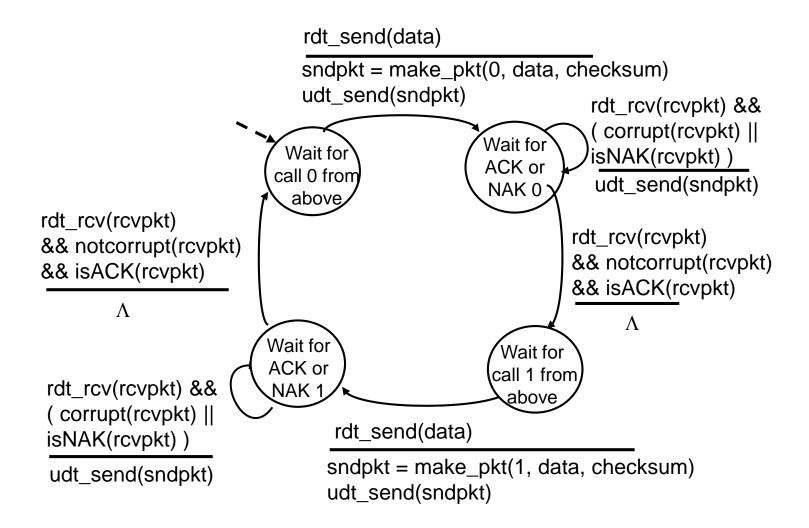
- sender doesn't know what happened at receiver!
- can't just retransmit:
 possible duplicate

handling duplicates:

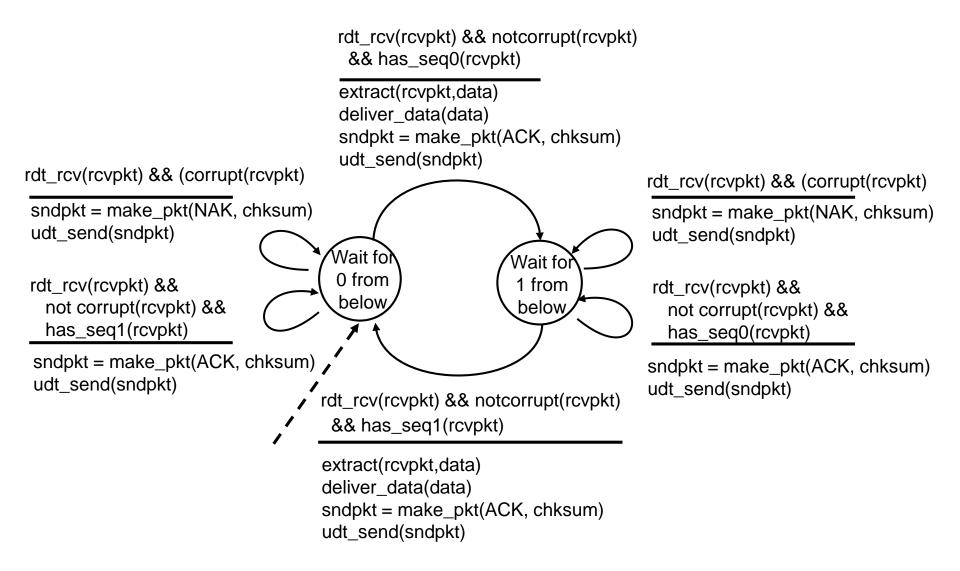
- sender retransmits current pkt if ACK/NAK corrupted
- sender adds sequence number to each pkt
- receiver discards (doesn't deliver up) duplicate pkt

stop and wait
 sender sends one packet,
 then waits for receiver
 response

rdt2.1: sender, handles garbled ACK/NAKs



rdt2.1: receiver



rdt2.1: summary

sender:

- seq # added to pkt
- two seq. #'s (0,1) will suffice. Why?
- must check if received ACK/NAK corrupted
- twice as many states
 - state must "remember" whether "expected" pkt should have seq # of 0 or 1

receiver:

- must check if received packet is duplicate
 - state indicates whether
 0 or I is expected pkt
 seq #
- note: receiver can not know if its last ACK/NAK received OK at sender



- Midterm on Chapters I and 2
 - Exclude Sec 1.7, 2.4, 2.7, and 2.8
- ☆ ~10% concepts addressed in class
- ~90% problems requiring math models
 - In addition to Assignments I and 2, Tutorials
 I to 4, reading from the book, make sure you
 have no doubts on the following problems:
 - Ch1: P2-P17, P19-P30, and P32
 - Ch2: PI, P7-PII, P20-P33
 - Note that our reference is the 5th edition