Transmission Control Protocol

Internet Architecture

Internet Architecture

- End-to-end semantics:
 - functions placed at low levels of a system may be redundant or of little value when compared with the cost of providing them at that low level

e.g., end-to-end caretaking A --- B --- C --- D --- E

J.H. Saltzer, D.P. Reed and D.D. Clark. End-to-end design argument in system design, ACM TOCS, Vol 2, Number 4, November 1984, pp.277-288.

Internet Architecture (Cont'd)

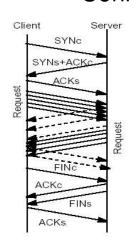
- Intermediate node (e.g., routers): packet forwarding, FIFO (first in first out), discard packets when overflow, stateless, know almost nothing about end-to-end sessions
 - → simple and robust
- End host: know almost nothing about network internals, manage all end-to-end session-related states
 - → complex and intelligent

Transmission Control Protocol

- TCP
 - a connection-oriented, end-to-end reliable, and stream-like transport protocol over the connectionless, unreliable, and datagrambased IP service

Transmission Control Protocol (TCP)

Connection-oriented

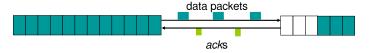


- connection establishment: associate endpoints
 - 3-way handshakes open bidirectional data channels
 - data only transferred after connection established (with bsd socket interface)
 - 2-way handshakes close data channels individually (i.e., graceful close. other termination forms exist)

	ТСР Н	eader
	Bit No	umber
	111111	111122222222233
0123456789	012345	6789012345678901
Source Port		Destination Port
	Sequence	Number
Ac	knowledgr	ment Number
Offset (Header Length) Reserved	Flags	Window
Checksum		Urgent Pointer
	Options (optional)
TC	P Header	Contents

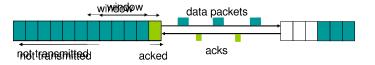
Error Control

- · error control: detect and recover packet errors
 - lost packet: accumulative acknowledgment from receiver, timeout at sender, retransmit if timeout occurs
 - duplicated packet: discard according to sequence number (note: the relation between window space and sequence space) at receiver, no further recovery needed



Flow Control

- flow control: coordinate endpoints
 - slide window based flow control (in sequence space)
 - receiver's acknowledgment advances window and adjust window size
 - sender can not exceed window-bottom plus window-size in sequence space



Error Control (Cont'd)

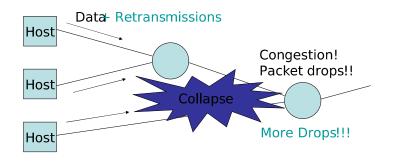
- corrupted packet: discard according to checksum (also include a pseudo IP header) at receiver, as a lost packet
- reordered packet: reorder according to sequence number at receiver, no further recovery needed

Congestion Control

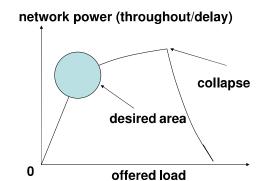
- congestion control: coordinate endpoints and network
 - was introduced in later 1980's
 - now standard in every TCP implementation
 - at the heart of TCP/IP research

Internet Congestion Collapse

 In the late 80s, the Internet suffered a series of congestion collapse

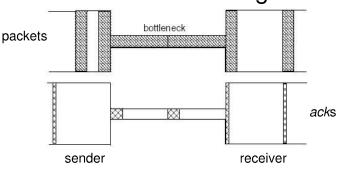


Congestion Control Objective



- load-power curve
 - low-load:significant gain
 - mid-load: negligible gain
 - high-load:negative gain (collapse)

ack self-clocking



In equilibrium: a new packet isn't put into the network until an old packet leaves.

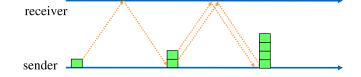
TCP Variants - TCP Tahoe

- cwnd (congestion window)
- · initial operation
 - cwnd = 1 mss (maximum segment size)
 - ssthresh (slow-start threshold)
 - max. outstanding dataswnd = min{cwnd, rwnd, sender buffer size}

Van Jacobson and Mike Karels. Congestion avoidance and control. ACM Computer Communication Review, 18(4):314--329, August 1990. *Revised version of his SIGCOMM '88 paper*.

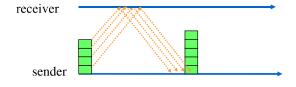
Slow-start

- slow-start: when cwnd < ssthresh
 - cwnd += 1 mss on a new ack
 - for ack-every-segment receiver $cwnd_{i+1} = cwnd_i * 2$
 - for delayed-ack receiver $cwnd_{i+1} \approx cwnd_i * 1.5$



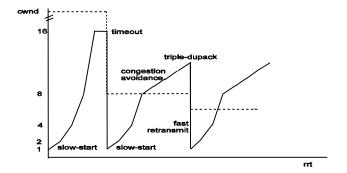
Congestion Avoidance

- congestion avoidance: when *cwnd* ≥ *ssthresh*
 - new ack, $cwnd += \frac{mss}{cwnd} mss$
 - for ack-every-segment receiver, $cwnd_{i+1} = cwnd_i + 1mss$
 - for delayed-ack receiver $cwnd_{i+1} \approx cwnd_i + \frac{mss}{2}$



Timeout

- packet is assumed lost when timeout occurs
- packet loss is assumed due to network congestion
- retransmit lost packet ssthresh=max{2 mss, \(\frac{\text{min}{cwnd,amount-of-in-flight-data}}{2}\)} cwnd=1 mss
- · followed by slow start



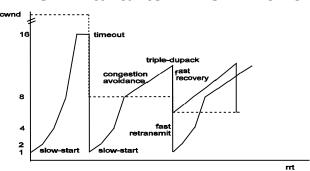
TCP New Reno

- partial acknowledgment: recover multiple packet losses
 - when entering fast recovery, record highest seqno ever sent as (record)
 - a newack not covers record is evaluated as a pack, retransmit the most unacknowledged packet
 - when *newack* covers *record*, exit fast recovery

Fast Retransmit

- retransmission after timeout is slow
- receiver returns duplicated acknowledgment on non-in-order packets (due to reordered or lost packets)
- sender assumes the most unacknowledged packet is lost if number of dupacks exceeds a certain threshold (3 in most popular TCP implementations)
- retransmit the most unacknowledged packet, adjust cwnd and ssthresh as a timeout occurs
- · followed by slow-start procedure

TCP Variants – TCP Reno



 Fast recovery: for triple-dupack, retransmit most unacknowledged packets, set ssthresh as fast retransmit, set cwnd = ssthresh

TCP SACK

- selective acknowledgment: recover multiple packet losses
 - receiver returns sack option and the first field specifies a hole in queue when a packet arrives
 - duplicates the first two sack fields in previous sack option
 - if a packet is reported in a hole three times, it is assumed lost and is arranged for retransmission

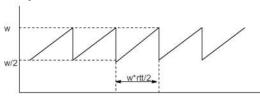
TCP Throughput

- · For saturated sender, TCP throughput (bytes/sec) can be approximated as the sender window size (bytes) over round-trip time (seconds).
- Assuming rwnd > > cwnd, the sender window size equals cwnd, which is determined according to the loss events.

TCP Throughput

• triple-dupack model

$$\frac{3w^2}{8} = \frac{1}{p}$$
, w: cwnd, p: loss rate



- limitation: small packet loss rate, periodically loss
- other models: consider delack and assume dropwindow-tail loss

25

TCP Throughput

- · model with timeout
 - BSD timeout backoff:

1, 2, 4, 8, 16, 32, 64, 64, ...

- correlated losses in rtt, independent between rtt packet

$$-\frac{1}{rtt\sqrt{\frac{2bp}{3}}+T_0\min(1,3\sqrt{\frac{3bp}{8}}p(1+32p^2))}, b: ack \text{ pattern}$$

- TCP-Friendly Rate Control (TFRC) protocol

26