Internet Architecture

Entities

- Hosts: end systems (e.g. laptops, smartphones)
- Communication links (e.g. radio satellite, fiber, copper)
- Packet switches (e.g. routers, switches)
- Network edge: Hosts
- Network Core: interconnected routers

Delays

- Store-and-forward: entire packet must arrive at a router before it can be transmitted on next link
- End-to-End delay: total time taken to send one packet from src to dest
 d_{node} = d_{prop} + d_{queue} + d_{trans} + d_{proc}
- Transmission delay = L/R
- Queuing delay: Total = n(n-1)(L/(2R))

Latency (s)	Bandwidth (bps)	Throughput (bps)
Time for 1 bit of data	Maximum number of	Number of bits
to travel from one	bits that can be	transmitted per unit
endpoint to another	transmitted per unit	time
	time	

- End-to-end throughput = transmission rate of bottleneck link
- Average and instantaneous throughput is independent of file size

Application Layer

- Resides in end-systems
- IP addresses: uniquely identifies host
- Port: identifies a process/service running on the host
- Socket: software interface to allow applications to send and receive messages

HTTP Protocol

- <u>Hypertext Transfer Protocol</u>
- Uses TCP, port 80
- Stateless

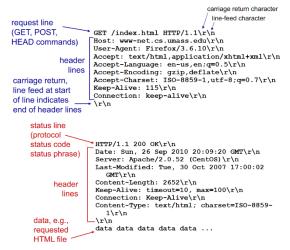
Non-Persistent HTTP

- Non-pipelined:
 - 1. Inititate TCP connection (1 RTT)
 - 2. GET HTML file (1 RTT + trans)
 - Connection closes
 - 4. Repeat for each *n* additional ojects
- Pipelined (m files at once):
 - 1. Initiate TCP connection (1 RTT)
 - 2. GET HTML file (1 RTT + trans)
 - Connection closes
 - 4. Repeat for $\lceil n/m \rceil$ times:
 - a. Initiate TCP connection (1 RTT)
 - GET m objects (1 RTT + m trans)
 - c. Connection closes

Persistent HTTP

- Non-pipelined:
 - 1. Inititate TCP connection (1 RTT)
 - 2. GET HTML file (1 RTT + trans)
 - 3. Repeat for *n* times:
 - 3.1. GET object (1 RTT + trans)
 - Connection closes
- Pipelined (*m* files at once):
- 1. Initiate TCP connection (1 RTT)
- 2. GET HTML file (1 RTT + trans)
- 3. Repeat for [\sfrac{n}{m}] times:
 - 3.1. GET m objects (1 RTT + m trans)
- 4. Connection closes

HTTP Request and Response



Cookies

- 1. Client sends HTTP request to server for the first time
- Server responds with cookie-ID in response header
- 3. Client accepts cookie?
 - 3.1. Yes: Cookie saved in client until TTL expires. Server creates entry in backend database
- 4. [If accepted] Client sends next HTTP request, with cookie-ID in request header
- [If accepted] Server stores and/or retrieves user-specific information from database using the cookie-ID and sends user-specific response

Web proxys

- 1. Client sends request to proxy server
- 2. Proxy server checks if response is cached:
 - 2.2. Cached: response with cached response. Done
 - 2.3. Step 3
- . Proxy server forwards original request to origin server
- Proxy server caches the response from origin server and responses back to client

- Conditional GET: Allows proxy server to query origin server to check if the cached response data has changed since it was last cached
 - 1. Client sends GET request to proxy server
 - Proxy server sends Conditional GET request to origin server with "Ifmodified-since: <date-in-cached-response>" in header
 - 3. Origin server responds with:
 - 3.1. 304 Not Modified if response data was modified after date-in-cached-response. Empty response body
 - 3.2. 200 OK if data was not modified
 - Proxy server updates the new response (if any) and responds back to client

DNS

- <u>D</u>omain <u>N</u>ame <u>S</u>ystem
- Maps host name to unique IP address
- Uses UDP, port **53**
- Local DNS servers do not belong to the global hierarchy of distributed servers
- Root DNS servers: top most in the hierarchy
- Top-level domain (TLD) servers: .com, .net, .edy etc.
- Authoritative DNS servers: organisations' servers e.g. .nus
- DNS resolution:
 - 1. Host makes request to Local DNS server
 - 2. Cached?
 - 2.2. Yes: Responds host. Done
 - 2.3. No: Step 3
 - 3. Local DNS server forwards query to root DNS server
 - Root DNS server responds local DNS server with the next lower level DNS server X it should query next (e.g. .com)
 - 5. Local DNS server queries DNS server X...
 - 6. Repeat iteratively until host name is resolved
- Local DNS servers process DNS queries in a recursive manner, where as the distributed servers do so in an iterative manner

Socket Programming

- UDP: SOCK DGRAM: TCP: SOCK STREAM
- TCP: client calls .connect() to establish handshake
- UDP: uses sendto() with destination addr and port as parameters
- TCP: uses send()
- Client's port is automatically chosen by OS; Server needs to bind to correct port with .bind()
- Client can call .bind() to manually choose which port to use

Transport Layer

- Provides process-to-process communication between hosts
- Resides in end-systems

UDP Protocol

- User Datagram Protocol
- Unreliable data transfer; connection-less
- Not full duplex
- UDP socket is identified by (dest IP, dest Port)

UDP header		
Source Port	Dest Port	
(2 bytes)	(2 bytes)	
Length	Checksum	
(2 bytes)	(2 bytes)	

- Both length and checksum includes header
- Checksum:
 - Sender: 1's complement addition with every 16-bit integers (i.e. includes carry-out, inverting bits after sum)
 - Receiver: sum every 16-bit integer, then add the sum to checksum. If all bits of result is 1, no error

Principles of Reliable Data Transfer

Utilisation: fraction of time sender busy sending

$$U = \frac{L/R}{RTT + L/R}$$

Stop-and-Wait Protocols

- Low utilisation

rdt1.0

- Assumptions: underlying channel is perfectly reliable
- Sender:
 - Sends packet
- Receiver:
 - Receives packet

rdt2.0 (Checksum, ACK/NAK)

- Assumptions: bits may be flipped
- Checksum: detect errors
- ACK/NAK: recover from errors
- Sender:
 - Sends packet
 - ACK/NAK is corrupted or NAK: retransmit
- Receiver:
 - Packet is not corrupted: Responds with ACK
 - Packet is corrupted: Responds with NAK
- Flaw: Does not deal with duplicate packets received at the receiver

rdt2.1 (Checksum, ACK/NAK, seg #)

- seq #: receiver can detect duplicates from retransmission by sender
- Sender:
 - Sends packet
 - ACK/NAK is corrupted or NAK: retransmit
- Receiver:
 - Packet is not corrupted: Responds with ACK
 - Packet is corrupted: Responds with NAK
 - Packet is a duplicate: Discards packet and responds with ACK

rdt2.2 (Checksum, ACK, seq #)

- Sender:
 - Sends packet
 - ACK is corrupted or duplicate: retransmit
- Receiver:
 - Responds with ACK # = last packet received
- Packet is a duplicate: Discards packet and responds with ACK # = last packet received

rdt3.0 (Checksum, ACK, seq #, Timer)

- Assumptions: bits may be flipped and packets can be lost
- Sender:
 - Sends packet
 - Timeout: retransmit
 - ACK is corrupted or duplicate: ignore, do nothing
- Receiver:
 - Responds with ACK # = last packet received
 - Packet is a duplicate: Responds with ACK # = last packet received
- May introduce packet reordering; solution: use larger range of sequence numbers

Pipelined Protocols

Go-Back-N

- Uses cumulative ACK
- Single timer; timer is for the oldest unACKed packet (in the window)
- Sender
 - Window will always contain only unACKed packets
 - ACK $n \rightarrow$ advance window base to n + 1, then sends packets newly added into the window
 - ACK is corrupted or duplicate: ignore, do nothing
 - Timeout: retransmit all packets in the window
- Receiver
 - Window size = 1
 - Responds with ACK # = last <u>correctly</u> received, <u>in-order</u> packet
- Packet is corrupted, duplicate or out-of-order: discard, responds with appropriate ACK

Selective Repeat

- Uses individual ACKs
- Uses individual timers
- Sender
 - Window can contain both unACKed and ACKed packets
 - Window base is always at the oldest unACKed packet
 - ACK base# → advance window base to # of oldest unACKed packet
 - Sends packets newly added into the window
 - ACK not in window → ignore, do nothing
 - ACK is corrupted or duplicate: ignore, do nothing
 - Timeout: retransmit single packet
 - Receiver
 - Responds with ACK # = # of packet, if uncorrupted
 - Packet is corrupted: discard, no ACK is sent

- Packet is out-of-order: buffer, then ACK
- Packet is in-order: delivers all buffered, in-order packets and advance window to # of next not-yet
- Packet is duplicate: discard, then ACK

TCP Protocol

- Transmission Control Protocol
- Reliable data transfer; connection-oriented
- TCP socket is identified by (src IP, src Port, dest IP, dest Port)
- Full duplex

	- 32	oits	
	source port #	dest port #	segment seq #: counting
ACK: seq # of next expected	sequence	number —	 bytes of data into bytestream
byte; A bit: this is an ACK	acknowledge	ment number	(not segments!)
length (of TCP header)	head not len used C E UAPRSF	receive window -	— flow control: # bytes
Internet checksum —	checksum	Urg data pointer	receiver willing to accept
C, E: congestion notification	options (vari	able length)	
TCP options	anni	ication	data sent by
RST, SYN, FIN: connection		ata —	application into
management	(variab	le length)	TCP socket

Data transfer

- MSS = maximum size of payload
- Seq # = ith position of the first byte in payload in full data; need not start at 0
- ACK # = expected next seq #
- Uses cumulative ACK
- Single timer; timer is for the oldest unACKed packet (in the window)
- Sender
 - Window will always contain only unACKed packets
 - ACK n → advance window base to n + 1, then sends packets newly added into the window
 - Timeout: retransmit single packet
- Receiver
 - Packet is out-of-order: buffer or discard, then respond with duplicate ACK
 - Packet is in-order: responds with appropriate ACK
- Packet is corrupted: discard, no ACK is sent
- Congestion control: timeout interval doubles after each timeout for each packet
- Fast retransmit: if sender receives 4 consecutive duplicate ACKs, retransmit the corresponding packet

Receiver	Action	
Arrival of in-order packet	Delayed ACK (500ms) and wait for	m
with expected seq #	next packet. If no packet, send ACK	Ш
Arrival of in-order packet	Immediately send single cumulative	
with expected seq #, the one	ACK, essentially ACK-ing both in-	
before has pending ACK	order packets with one ACK	
Arrival of out-of-order	Immediately send duplicate ACK	
packet higher than expected		Ш
seq #		
Arrival of out-of-order	Immediately send ACK	
packet higher than expected		
seq # that		
partially/completely fills gap		

Timeout Estimation

$$\label{eq:timeoutInterval} \begin{split} & \textit{TimeoutInterval} = \textit{EstimatedRTT} \ + \ 4 \times \textit{DevRTT} \\ & \textit{EstimatedRTT} \ = \ (1-\alpha) \times \textit{EstimatedRTT} \ + \ \alpha \times \textit{SampleRTT} \\ & \textit{DerivedRTT} \ = \ (1-\beta) \times \textit{DevRTT} \ + \ \beta \times |\textit{SampleRTT} \ - \textit{EstimatedRTT}| \\ & - \text{Typically, } \alpha = 0.125, \beta = 0.25 \end{split}$$

Three-way Handshake

- 1. Client sends SYN segment with SYN = 1 (seg # = client isn)
- Server sends SYNACK segment with SYN = 1 (seq # = server_isn, ACK # = client_isn)
- Client sends segment with SYN = 0 (seq # = client_isn + 1, ACK # = server_isn + 1)

Connection Termination

- 1. Client sends segment with FIN = 1 (seq # = x)
- 2. Server reponds with segment with ACKbit = 1 (ACK # = x + 1)
- 3. Server sends segment with FIN = 1 (seq # = y)
- 4. Client responds with segment with ACKbit = 1 (ACK # = y + 1)
- 5. Connection is terminated
- Either client or server can initiate the connection termination
- Once a FIN segment is send, the host can no longer send application data except for retransmitting segments sent before the FIN segment

Flow Control

- Receiver
 - Maintains LastByteRcvd, LastByteRead, RcvBuffer
 - Amount of data in buffer = LastByteRcvd LastByteRead
 - Attaches rwnd = RcvBuffer (LastByteRcvd LastByteRead) to ACK packet
- Sender
 - Maintains rwnd, LastByteSent, LastByteAcked
 - Amount of data sent = LastByteSent LastByteAcked
 - Ensures that LastByteSent LastByteAcked ≤ rwnd whenever it wants to send a segment
- When rwnd = 0, sender continues to send 1 byte to receiver
 - Receive buffer is full: byte is dropped
 - Receive buffer is not full: Receiver sends ACK segment, letting sender know that buffer is not full

Network Layer

- Provides host-to-host communication
- Resides in network core
- Forwarding: move packets to appropriate router output port
- Routing: determine route taken from source to dest

IP address

- Per interface
- 32-bits
- a.b.c.d/x, x = subnet prefix
- Subnet mask: x MSB 1s, 32-x 0s

- 0.0.0.0/8: non-routable meta-address
- 255.255.255.255/32: broadcast address

Subnets

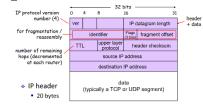
- Hosts can connect to one another without a router
- Hosts have same network prefix

DHCP Protocol

- <u>Dynamic Host Configuration Protocol</u>
- Dynamically obtain IP address when host joins a network
- Uses UDP, port 68 for client, port 67 for server
- ID identifies the client
- 1. DHCP discover: client sends
 - <src: 0.0.0.0, 68, dest: 255.255.255,255, 67, yiaddr: 0.0.0.0, ID>
- 2. DHCP offer: server responds
 - <src: X.X.X.X, 68, dest: 255.255.255, 67, yiaddr: Y.Y.Y.Y, ID, lease time>
- 3. DHCP request: client sends
 - <src: 0.0.0.0, 68, dest: 255.255.255,255, 67, yiaddr: Y.Y.Y.Y, ID + 1, lease time>
- 4. DHCP ACK: server responds
 - <src: X.X.X.X, 68, dest: 255.255.255,255, 67, yiaddr: Y.Y.Y.Y, ID + 1,
 lease time>
- DHCP discover and offer are not mandatory
- Also returns address of first-hop router, name and IP of DNS server and network mask
- Client chooses any offer should there be multiple servers reponding

IP Protocol

Internet Protocol



Fragmentation

- Max Fragment size = MTU of link-level frame
- Max payload size in fragment = MTU 20 bytes
- # fragments = [original size / MTU]
- Flag: 0 if is the last fragment, else 1
- Offset: Units of 8-bytes
- · ID: ID of original datagram

Hierarchical Addressing

- Route aggregation: combine multiple IP addresses into a singe route advertisement → reduces size of forwarding table
- Routing table stores < Dest IP addr range, Next Hop> entries
- Longest prefix-matching

Intra-Autonomous Systems Routing

- Autonomous system: a group of routers under a common administrative domain
- Finds path between two routers within AS
- Distance Vector Routing Algorithm
 - Each router X knows the cost C_{X,Y} of one hop
 - Each router X stores a routing table containing distance vectors Dx(W) to all other routers in the AS network
 - Initial $D_X(W) = C_{X,W}$ if W is one-hop away, else $D_X(W) = \infty$
 - When Dx(W) changes in router X, it updates its distance vectors then notifies its neighbours
 - Neighbours Y update their distance vectors then notifies their neighbours
- Routing Information Protocol (RIP): DV with $C_{X,Y} = \#$ hops from $X \rightarrow Y$
 - Uses UDP, port 520
 - Routing table exchanged every 30s
 - No update from neighbour for 3 minutes → delete entry

ICMP Protocol

- Internet Control Message Protocol
- For error reporting or router signaling
- ICMP header: Type, Code, Checksum
- ping
 - Type = 8, Code = 0
 - Checks if connection to a remote host is successful
- TTL expired: Type = 11, Code = 0
- Traceroute
 - Returns route taken by a packer from one host to another by setting TTL = # hops

NAT

- Network Address Translation
- 1. LAN host (private IP, port) sends datagram to NAT router
- NAT router modifies IP src IP and port in header: (private IP, port) → (NAT IP, unique port)
- NAT saves (NAT IP, unique port) → (private IP, port) mapping in its NAT translation table
- 4. NAT sends modified datagram to server in WAN
- Server responds back with datagram containing dest IP and port = (NAT IP, unique port)
- . NAT translates (NAT IP, unique port) → (private IP, port)
- 7. NAT forwards datagram to LAN host

Link Layer

- Resides in NIC card
- Involved adjacent nodes over a single link

Error Detection & Correction

1-D Parity Checking

- Detects all odd number of single bit errors, but cannot correct
- Poor performance with bursts or errors

2-D Parity Checking

- Detects all single or 2 bit errors
- Corrects all single bit errors

Cyclic Redundancy Check (CRC)

- Length of CRC = r
- Length of generator G = r + 1
- Sender:
 - 1. Append *r* number of 0s to data D
 - 2. Divide *D* by *G* (XOR)
 - 3. Remainder = CRC
- Receiver:
 - 1. Divide data D
 - 2. If remainder = 0, no errors
- Detects all <u>odd</u> number of single bit errors as well as all burst (consecutive bit) errors of ≤ r bits
- Detects burst errors of > r bits with $P = 1 0.5^r$

Multiple Access Protocols

- Ideal properties:
 - 1. Collision free
 - 2. Efficient: a transmitting node should transmit at max capacity
 - 3. Fairness: when n nodes want to transmit, each node transmits at a rate of R/n
 - 4. Fully decentralised: no external nodes

Channel Partitioning Protocols

Time Division Multiple Access (TDMA)

Collision Free	Efficient	Fairness	Decentralised
✓		√	\

- Divide into time frames and divide each time frame by # nodes
- Each node gets fixed length time slot to transmit at max throughput = R/n

Frequency Division Multiple Access (TDMA)

Collision Free	Efficient	Fairness	Decentralised
✓		✓	✓

 Divide channel spectrum into n frequency bands and each node transmits at one band at max throughput = R/n

Taking Turns Protocols

Polling Protocol

Collision Free	Efficient	Fairness	Decentralised
√	(polling overhead)	√	

 Master node polls which node to transmit next (and how many frames to transmit) in <u>round-robin</u> fashion

Token Passing Protocol

Collision Free	Efficient	Fairness	Decentralised
✓	(token passing overhead)	✓	✓

- Token (special frame) passed from one node to the next sequentially
- Only the node holding onto the token can transmit X # of frames
- If a node has nothing to transmit, it node immediately passes the token on to the next node
- Token loss possible → need to implement token recovery

Random Access Protocols

Slotted ALOHA

Collision Free	Efficient	Fairness	Decentralised
	✓	✓	✓

- Divide time into equal slots (of length ≥ time to transmit 1 frame);
 each node attempts to transmit at the beginning of a slot
- Collision: with probability p, transmit at next slot
- Efficiency ≈ 37%

Pure Division Multiple Access (TDMA)

Collision Free	Efficient	Fairness	Decentralised
	√	√	✓

- Nodes attempt to transmit at any given time
- Collision: wait for <u>1 frame transmission time</u>, then with probability p, transmit at next slot
- Efficiency ≈ 18%

Carrier Sense Multiple Access (CSMA)

Collision Free	Efficient	Fairness	Decentralised
	✓	✓	✓

- If channel is busy: wait for short while before sensing again
- If channel is idle: transmit
- Collision: Possible due to propagation delay; does not abort transmission

Carrier Sense Multiple Access with Collision Detection (CSMA/CD)

			1 - 1 -
Collision Free	Efficient	Fairness	Decentralised
	✓	✓	✓

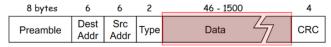
- If channel is busy: wait for short while before sensing again
- If channel is idle: transmit
- Collision: Possible due to propagation delay; <u>abort</u> transmission and retransmit after a <u>random</u> delay
- Exponential backoff: For each frame,
 - After the nth collision, next delay is random $K \in \{1, 2, ..., 2^n-1\}$ time units, up to n = 10
 - Time unit = 512 bit time (time taken to transmit 512 bits)
 - Actual delay = 512K/R
- Ethernet minimum frame size = 64 bytes to ensure d_{trans} > d_{prop} and collisions are detected

Ethernet

MAC Address

- Burnt into NIC card (i.e. fixed and globally unique)
- A host can have multiple MAC addresses as long as it has multiple NIC
- 48-bit hexadecimal e.g. 5C-F9-DD-E8-E3-D2
- Broadcast address = FF-FF-FF-FF-FF
- LAN: network that interconnects hosts within geographical area

Ethernet Frame



Network Layer Datagram

- src & dest addr: NIC only delivers packets up if dest MAC of received packet is broadcast address or own address
- Preamble: 7 bytes of 10101010 (AA) and 1 byte of 10101011 (AB) or "start-of-frame"
 - Synchronise sender and receiver clock rates
- Type: indicates the higher layer protocol (e.g. IP)

Physical Topology

- Bus:
 - Broadcast LAN, all devices connect to it
 - Bus is single point of failure, hard to troubleshoot and slow due to collisions

Star:

Hub	Switch
Physical layer device	Link layer device
Acts on individual bits	Acts on frames (store-and-
	forward)
Collisions possible	Simultaneous transmissions
	without collisions
Bits are copied, boosted and	Frames only transmitted to one
broadcasted to all interfaces	interface
Cheap	Expensive

Ethernet Switch

- Uses CSMA/CD
- Transparent, plug-and-play (self-learning)
- Buffers frames
- Frame filtering/forwarding (Host A → Host B, A ≠ B)
 - 1. Switch receives frame from interface X occupied by Host A
 - 2. Switch stores <MAC_A, Interface_X, TTL> in switch table
 - 3. Does switch table contain MAC B?
 - Yes: Forward to table[MAC B].Interface
 - No: Broadcast to all interfaces
- If src MAC = dest MAC, frame is filtered

ARP Protocol

- Address Resolution Protocol
- Maps IP address to MAC address
- ARP tables reside in each interface

Same Subnet (Host A → Host B)

- 1. Host A's ARP table contains IP_B?
 - Yes: Transmit ethernet frame with dest MAC = MAC_B
 - No: Transmit ARP request packet with dest MAC = FF-FF-FF-FF-FF
- 2. Switch receives frame from Host A
 - Switch keeps a record <MAC A, Interface X, TTL> if there isn't one
 - Not ARP request?
 - o Switch table contains MAC B?
 - Yes: Forward frame to table[MAC_B].Interface. DONE
 - No: Broadcast frame to all interfaces. DONE
 - ARP request?
 - o Broadcast frame to all interfaces
- 3. Host B receives the frame and responds with its MAC address
 - Host B keeps a record of <IP_A, MAC_A> in its ARP table if there isn't
 one
- 4. Switch receives frame from Host B
 - Switch keeps a record <MAC_B, Interface_Y, TTL> if there isn't one
- 5. Switch forwards frame to Interface_X (Host A)
- Host A keeps a record of <IP_B, MAC_B, TTL>. Repeat from 1 with IP datagram instead of ARP request

Same Subnets (Host A → Host B)

- Host A creates and transmits ethernet frame with src MAC = MAC_A and dest MAC = MAC_Router_Interface_A to router
- Router has two interfaces: one resides in Host A's subnet (Interface_A) and the other in Host B's subnet (Interface_B)
- Interface_A of router receives the frame, extracts the IP datagram (to see Host B's address)
- 4. Router checks Interface B's ARP table for Host B's MAC address
 - If it does not exist, transmit ARP packet to switch in subnet B
- Router modifies the frame and forwards the frame with src MAC = MAC_Router_Interface_B and dest = MAC_B to Host B