Network Layer Overview

Services and Protocols

- To transport **segments** from the **sending host**, to the **receiving host** the following happens:
 - 1. The sender encapsulates segments into datagrams and passes them to the link layer.
 - 2. The receiver delivers segments to the transport layer protocol.
 - 3. A router is a piece of network hardware than manages traffic between networks.
 - Routers work by examining the headers in IP datagrams (Packets), and move the datagrams from input ports to output ports; with the goal of transfering datagrams along the end-end path.
 - Routers work a the **Network Layer (Layer 3)**, and also use layers 1 and 2 to facilitate the data transfer.
 - Routers us Internet Protocol Addresses (IP Address) to identify networks / hosts.

Key Network-Layer Functions

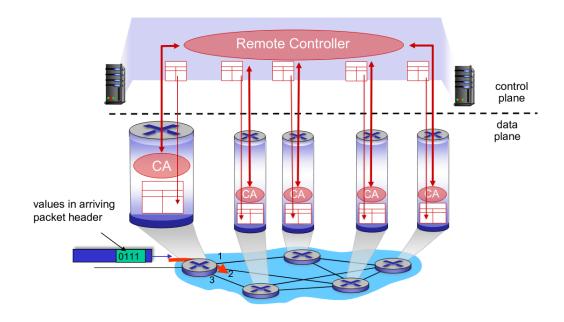
- One key network-layer function is **forwarding**, **forwarding** involves **moving packets** from a **rounter's input link** to the appropriate **output link**.
- Another key network-layer function is **routing**, **routing** involves **determining the route taken by packets** from the **source** to the **destination**.
 - There are many routing algorithms that can be used the achieve this.

The Data Plane vs The Control Plane

- The data plane is a local, per-router function that determines how packets arriving on a router's input port is forwarded to router's output port.
- The control plane is a network-wide function, that determines how packets are routed amongst routers along end-end paths from source host to destination host.
 - There are two control-plane approaches:
 - 1. Traditional routing algorithms that are implemented in routers.
 - 2. Software-defined networking (SDN) that is implemented in remote servers.

Per-Router Control PLane Software-Defined Networking (SDN) Control Plane

- Per-Router control plane consits of a routing algorithm in every router that interacts with the control plane. Each router determines where to route the packets.
- SDN is composed of remote controller computers, that install forwarding tables in router. The routers then use these tables to forwards packets.



Network Service Models

• Internet service models:

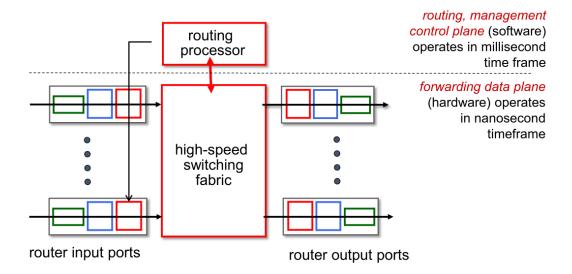
Network Architecture		Service Model	Quality of Service (QoS) Guarantees ?				
			Bandwidth	Loss	Order	Timing	
	Internet	best effort	none	no	no	no	
	ATM	Constant Bit Rate	Constant rate	yes	yes	yes	
	ATM	Available Bit Rate	Guaranteed min	no	yes	no	
	Internet	Intserv Guaranteed (RFC 1633)	yes	yes	yes	yes	
	Internet	Diffserv (RFC 2475)	possible	possibly	possibly	no	

• Though the **best effor service model** may not provide any guarantees, it allowed the internet to be widely deployed, and adopted.

Router Architecture Overview

Routers

- A router is a networking device that forwards and router data packets between networks.
- Routers have input ports and output ports, to receive and forward packets respectively.



- The green boxes represents the **physical layer**, the blue boxes represent the **link layer**, and the red boxes represent the **network layer**.
- The first red box continas a queue of packets that need to be forwarded, and the lookup table, that map headers to ports.
- **Destination-based forwarding** is forwarding based only on the **destination IP Address** (traditional).
- Generalized forwarding is forwarding based on any set of header field values.
- The following is an example of a lookup table:

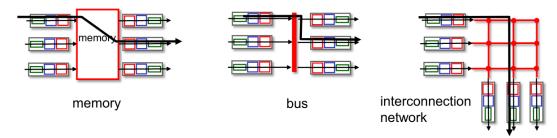
Destination Address Range	Link Interface	
11001000 00010111 000 <mark>10000 00000000</mark> through	0	
11001000 00010111 000 <mark>10111 11111111</mark>		
11001000 00010111 000 <mark>11000 00000000</mark> through	1	
11001000 00010111 000 <mark>11000 11111111</mark>		
11001000 00010111 000 <mark>11001 00000000</mark> through	2	
11001000 00010111 00011111 11111111		
otherwise	3	

 To determine which interface a an IP address should be mapped to, you see what address range has the longest prefix that matches the IP address of the packet that is being router.

Switching Fabrics

• Switching fabrics are responsible for transfering the packet from the input link to the appropriate output link.

- The switching rate is the rate at which packets can be transferred from inputs to outputs.
- There are three main types of switching fabrics:



- With **memory switching**, the packets are copied to the **system's memory**. This limits the switching rate to the **memories bandwidth**. This type of switching is directly under the control of the **CPU**.
- With **bus switching**, the packets are delivered from the **input port's memory**, to the **output port's memory** directly. The switching speed is limited to the **speed of the bus** (which is much faster than memory switching).
- With interconnnection network switching, is similar to the bus, except we can transfer several packets in parallel.
- We can have **seveal planes** of **interconnection network switching** that run in parallel to allow for scaling.

Input Port Functions

- There are three functions of an **input port**:
 - 1. Receives the bits.
 - 2. Interperates the bits.
 - 3. Forward the packet to the switching fabric.
- There are two types of **forwarding** in **decentralized switching**:
 - Destination-based forwarding Packets are forwarded based on their destination IP Address.
 - 2. **Generalized forwarding** Packets are forwarded based on any set of header field values.
- If the switching fabric is slower than the input ports combined, then queueing may occur at the input queues. The queueing may lead to delays, and packet loss.
- Head-of-Line (HOL) blocking is when the packet at the front of the queue prevents the queue from moving (due to congestion).

Output Port Functions

- There are three functions of an **output port**:
 - 1. Receive the packet from the switching fabric.
 - 2. Send the bits.
 - 3. Send a line termination.
- Buffering is required when packets arrive from the switching fabric faster than the link's transmission rate.

- The drop policy is a policy to determine which packets to drop if the buffer is full.
- Congestion occurs when the output link is slower than the output ports.

Determining Buffer Sizes

- The RFC 3439 rule of thumb is that the average buffer size should be equal to RTT × LinkCapacity.
- There is such a thing as **too much buffering**; this can increase delays (particularly in home routers).
 - Long RTTs lead to poor performance in real-time allocations.
- When the **buffer is full**, there are two ways to determine the **packets to drop** when more arrive:
 - 1. **Tail drop** involves dropping the arriving packets.
 - 2. **Priority drop** involves dropping packets on a priority basis.

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Packet Scheduling

- There are several ways to determine the way packets are scheduled:
 - 1. First Come First Serve (FCFS) Scheduling schedules packets in the order they arrive.
 - 2. Priority Scheduling schedules packets based on their classification (classifications can be determined by header fields); all higher classified queues get send first.
 - 3. Round Robin (RR) Scheduling schedules packets based on their classification, but cycles between the different classification queues.
 - 4. Weighted Fair Queueing (WFQ) is a round robin scheduling algorithm, that assigns each class a weight, and the weight determines the amount of time is spent on each queue.

The Internet Protocol

Internet Protocol Datagrams

32 bits									
sion Header length Type of service		Datagram length (bytes)							
16-bit Identifier			13-bit Fragmentation offset						
o-live	Upper-layer protocol	Header checksum							
32-bit Source IP address									
32-bit Destination IP address									
Options (if any)									
Data									
	length	Header length Type of service 16-bit Identifier O-live Upper-layer protocol 32-bit Source 32-bit Destina	Header length Type of service If 16-bit Identifier Flags To-live Upper-layer protocol 32-bit Source IP add 32-bit Destination IP Source IP add Options (if any						

• The maximum length of a datagram is 65536 bytes. However the typical size is around 1500 bytes or less.

IP Address

- An IP Address is a 32-bit identifier associate with each network host or router interface.
 - An interface is a connection between a host/router and a physical link.
 - Routers typically have multiple interfaces, and hosts typically have one or two interfaces.

Subnetworks

Subnetworks (Subnets)

- A subnetwork is a local partition of an Internet Protocol network.
- Subnets are typically defined as a group of device interfaces that can physically reach each other without passinging through an intervening router.
- Devices in the same subnet have common high order bits in their IP Addresses, and the low order bits differ between hosts.
 - For example, the following addresses are an example of a subnet: 1.0.23.1, 1.0.23.5, 1.0.23.10
- A subnet mask is a 32-bit number created by setting host bits to 0, and network bits to 1. All bits set to 0 can change, and the addresses will still be a part of the subnet.

- There are three main classes of networks:
 - 1. Class A networks have a subnet mask of 255.0.0.0.
 - 2. Class B networks have a subnet mask of 255.255.0.0.
 - 3. Class C networks have a subnet mask of 255.255.255.0.
- There is also a **Class D and Class E**, but they are reserved for special purposes, and research. They are not available for network hosts.

Gateways

- Each inteface on a router has a unique IP Address.
- Routers that provide an interface between two distinct subnetworks are known as a gateway.
 - Gateways are a router that interfaces multiple subnets, hence why each interface gets a unique IP Address that must be within the subnet it is interfacing with.
 - When at least two gateways are directly connected, there is a new subnet between them.

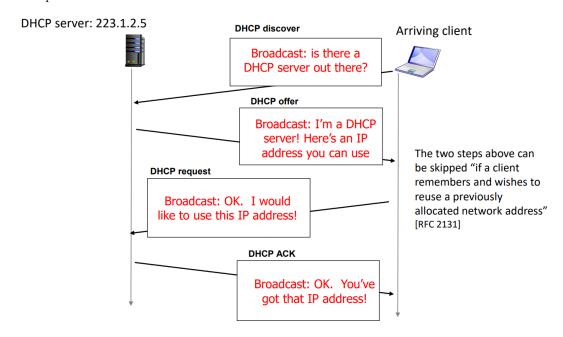
IP Adressing — CIDR

- Classless InterDomain Routing (CIDR) (aka supernetting) is a method of assigning Internet Protocol (IP) adresses.
- CIDR consists of the following address format: a.b.c.d/x where a, b, c, d are n-bit numbers numbers, and x is the number of bits in the subnet portion of the address.
 - An example would be **200.23.16.0/23**, the first 23 bits identify the subnetwork, and the last 9 bits identify the host within the subnetwork.
- There are to ways a **host** gets an **IP Address assigned**:
 - 1. A Hard-coded address that is assigned by a sysadmin in a configuration file.
 - 2. The Dynamic Host Configuration Protocol (DHCP) which allows devices the get IP Adress assignments dynamically from a server.
- DHCP is particularly useful for devices connecting to routers over Wi-Fi.

Dynamic Host Configuration Protocol

- The goal of **DHCP** is to allow **hosts** to **dynamically obtain IP Adresses** from a **network server** when it "joins" the network.
- DHCP "leases" out IP Adresses to devices on the network for a specific period of time. This allows for addresses to be reused if a device leaves the network.
- The process is **seamless** so hosts can easily come and go.
- The following is how a device typically obtains an IP Address over DHCP:
 - 1. The **host** broadcasts a **DHCP** discover message (only if the host does not know the address of the DHCP server).
 - 2. The **DHCP server** responds with a **DHCP offer** message (only if the host sends a discover message).

- 3. The **host** requests the **IP Address** that the **server** offered with a **DHCP request** message.
- 4. The DHCP server sends a DHCP ack message to the address that has been leased to the host.
- Typically, the **DHCP server** will be **co-located in the router**, serving all **subnets** the **router is attached to**.
- When **DHCP** messages are sent, the **server** typically uses port **67**, and the host typically uses port **68**.



- DHCP can return more than just an IP Address, it can also return:
 - 1. The address of the first-hop router for a client.
 - 2. The name and IP of a DNS server.
 - 3. The **network mask** (including network vs host portion of an address).

Internet Service Providers

- The Internet Corporation for Assigned Names and Numbers (ICANN) is responsible for assigning IP address blocks to Internet Service Providers (ISPs).
- The **ISPs** can then assign them to their clients.

Network Address Translation (NAT)

- All devices in a local network share one public IPv4 address as far as the outside world is concerned (the address of the first-hop router).
- Datagrams with a source or destination within this network have 10.0.0.0/24 or 192.168.0.0/24 as the source and destination address (usually).
- All datagrams that are leaaving the local network have the same source NAT IP Address, but different source port numbers.
- All devices in the local network have a private 32-bit IP Address that can only be used in the local network.

- The advantage that is provided by allowing **one public IP Address** to map to **several private IP Addresses** is:
 - 1. The ISP only needs to assigne one address to the customer for all devices on the local network.
 - 2. You can change the address of hosts in a local network without notifying the outside world.
 - You can change ISP providers without changing addresses of devices in the local network.
 - 4. Enhanced security; devices inside the local network are not directly addressable or visible by the outside world, instead they are proxied.
- In 2011, ICANN allocated the last remaining available IPv4 addresses to an ISP, public and private IP Adresses allow us to remain on IPv4 for longer, because the same addresses are reused in several local networks.

NAT Implementation

- The NAT router must transparently do the following:
 - Outgoing Datagrams The source IP and port of every datagram must be replaced
 by the NAT IP and a new port which identifies the host in the local network. Remote
 hosts will respond using the new NAT IP address and port. The mapping will be placed
 in a NAT table.
 - 2. **Incoming Datagrams** The destination IP and port of every datagram must be replaced with the with the original local IP and port, which were previously stored in a NAT table.

The NAT Controversy

- Nat has been controversial:
 - Routers should only process up to layer 3 (ports are layer 4).
 - The address shortage should be solved by IPv6, not NAT.
 - It Violates the end-to-end argument (port # manipulation).
 - It doesn't allow clients to connect to servers behind NAT.
- Even with all of th controversy, NAT is here to stay. It is extensively used in home and institutional networks, and 4G/5G cellular networks.

Middlebox Functions

- A Middlebox is an intermediary box that performs functions apart from the noraml, standard functions of an IP router on the data path between a source host and a destination host.
- Some example of middleboxes are firewalls, NAT, load balancers, CDN cache, etc.
- Initially, middleboxes were proprietary hardware solutions.
- Over time, things moved towards **open solutions** with **programmable local actions** via match+action.
- Software-Defined Networking (SDN) allows for logically centralized control and configuration management, often in the "cloud".

Internet Protocol version 6 (IPv6)

IPv6 Format

- The **initial motivation** was the fact that **IPv4 addresses** would be (at the time they were not completely allocated, they are now) **completely allocated**.
- Unlike IPv4, the IPv6 format does not have a checksum, options, or fragmentation.
- A major problem is that not all routers can be upgraded simultaneously.
- To allow routers that only support IPv4, to handle IPv6 datagrams, we can use tunneling: The IPv6 datagram is carried as a payload in an IPv4 datagram. This is essentially, a packet within a packet.
- Tunneling is used extensively in other contexts such as 4G/5G.

