ECE560 Computer and Information Security

Fall 2023

Networking Overview

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Some slides adapted from Brian Rogers (Duke)

Network fundamentals

This course isn't a networking course, so we'll just hit the highlights

- We want to hook computers together
- There's a near-infinite number of ways to do this
- We'll skip the theory and show you the common case
- Average person's networking understanding

"Gateway where I get cat pics"

Network organization

Connectivity in the Internet

- A point-to-point mesh?
- Clearly not sustainable for large networks
	- \blacksquare N² links required
	- Add new endpoint: new link added to all existing endpoints

Network Structure

- Need to share infrastructure!
- Routers and switches (intermediate nodes) allow sharing

Internet Backbone

- From Wikipedia:
- Due to sharing, we get a structure that looks like this
- Localized "stars" connected to others

Usage Models

- Network endpoints run application programs
	- Web browser, email client, ssh, etc.
- Client / Server model
	- Client endpoints requests a service from a server
	- E.g. client / server web page service
- Peer-to-peer (P2P)
	- Direct client communication (e.g. Skype, BitTorrent)

Packet Switched Routers

- Multiplex w/ queue(s) in the router
- Demultiplex with packet header info:
	- **Destination endpoint**

Managing Complexity

- Very large number of computers
- Incredible variety of technologies
	- Each with very different constraints
- No single administrative entity
- Evolving demands, protocols, applications
	- Each with very different requirements!
- How do we make sense of all this?

Networking layers

Layering

• We see layers of abstraction

- Separation of concerns
	- Break problem into separate parts
	- Solve each one independently
	- Tie together through common interfaces: abstraction
	- Encapsulate data from layer above inside data from layer below
	- Allow independent evolution

Layering

Layering done wrong invites security vulnerabilities!

- Layering is a form of modularity; modularity is good *IF and ONLY IF* you don't make any dangerous assumptions!
- Networking stack is good, common, but has had lots of vulnerabilities
	- Many vulnerabilities of the form "layer X makes an implicit assumption about data from layer Y"
	- Example: Receiving a packet with an Ethernet frame size in conflict with TCP packet size -> Buggy network code segfaults
- **Rule of thumb**:

Be strict in what you send and check carefully what you receive

OSI Reference Model

TCP/IP Model

Layer 1 & 2

- Layer 1: Physical Layer
	- Encoding of bits to send over a single physical link

Examples: Ethernet, 802.11 WiFi (the part of the spec that says how to send bits)

- Layer 2: Link Layer
	- Framing and transmission of a collection of bits into individual messages sent across a single subnetwork (one physical topology)
	- Provides local addressing (MAC)
	- May involve multiple physical links
	- Often the technology supports broadcast: every "node" connected to the subnet receives

Examples: Ethernet, 802.11 WiFi (the part of the spec that how to send packets to a host on this network)

Ethernet/WiFi and MAC addresses

- Each network interface has a **MAC address** ("Media Access Control"): a 48-bit value burned into network card; globally unique
	- First 3 bytes tell the manufacturer (OUI: Organizationally Unique Identifier)
	- \blacksquare Last 3 bytes are made to be unique by that manufacturer
- Usually written as colon-delimited hex: BC:5F:F4:2B:E9:68
- Only meaningful on a single **local area network** (wired or wireless)
- Not transmitted across internet

Layer 1/2 demo: ARP

- Address Resolution Protocol (ARP): how we figure out the layer 2 address (MAC address) for a given layer 3 address (IP address)
	- Can inquire to see known MAC addresses
	- Can use OUI (first 3 bytes) to check manufacturer of devices!

Layer 3

- Bridges multiple "subnets" to provide end-to-end connectivity between nodes
- Provides global addressing (IP addresses)
- Only provides best-effort delivery of data
	- No retransmissions, etc.

28

Trace complete.

ms

 25 ms

29

 $\textsf{sc@FREEMAN}\sim\textsf{\$}$

 24 ms

 24 ms

 $rsv-1$

• Works across different link technologies

med out.

2-3-72-254.oit.duke.edu [152.3.72.254]

IPv4 addresses

- IPv4 address is 32-bit address that is (*theroetically*) globally unique; identifies interface on the internet.
- Written as "dotted decimal" of the four bytes, e.g. "141.9.68.24".
	- So each number ("octet") can be 0-255.
- Subnets
	- An address can have its bits divided into network and host.
	- \blacksquare We describe a network in dotted decimal with a suffix saying how many bits are in the network part, e.g.: 181.41.0.0/18 – this is a **subnet**.
	- A mask of one bits covering the network portion is called the **netmask**; for 181.41.0.0/18, the netmask would be 255.255.192.0
	- The number of hosts that fit in a subnet is $2^{32-n} 2$ (Minus two is because the all-zeroes host and all-ones host are special)
	- IP address assignment is hierarchical: Countries get IP ranges and assign to registrars who then divide them among customers (ISPs, companies, etc.).
	- The country of Aruba has 181.41.0.0/18 and a few others. For a long time, IBM had 9.0.0.0/8.

Modern caveats (1)

- Some IP addresses are special:
	- **Example 21 Loopback:** 127.0.0.1 always refers the machine you're on (actually it's all of 127.0.0.0/8)
	- **Private**: 10.0.0.0/8, 172.16.0.0/12, 192.168.0.0/16 not allowed on internet
	- **Link-local**: 169.254.0.0/16 auto-assigned when no network services are up
	- Others (see IANA IPv4 Special-Purpose Address Registry)
- We're running out of 32-bit IP addresses, so **NAT (Network Address Translation)** was invented
	- Have just one "real" public IP address at network boundary, assign private IP addresses internally and translate at border
	- $Extremely common real direct internet connections are rare$ (this is good, as NAT doubles as a firewall)

Modern caveats (2)

- IP-to-interface mapping is actually more flexible:
	- For performance/reliability, an IP may be span multiple interfaces
	- For manageability reasons, an interface may have >1 IP address
- All of the above refers to IP version 4 ("**IPv4**")
- **IPv6** is being deployed now (and has been for the past 20 years)
	- IPv6 addresses are 128 bits (16 bytes) instead of 32 bits (4 bytes)
	- \blacksquare Written as colon-delimited 16-bit hex words:

Looking at real configs: Windows

Looking at real configs: Linux

MAC address

Subnet mask (show this is a /24 network) • IPv4 address (NAT routed private IP) • IPv6 address (link local – not routed to internet in this config) • DNS server: IP address we look up names with (my router does this too in this case) • Gateway: IP address we sent stuff to go get to the internet (NAT router in this case)

Layer 4

- End-to-end communication between processes
- Different types of services provided:
	- UDP: unreliable datagrams
	- TCP: reliable byte stream
- "Reliable" = keeps track of what data were received properly and retransmits as necessary
- This is the layer that applications talk with

Connectionless vs. Connection

- Connectionless transport layer
	- Very similar to plain layer 4 (IP)
	- Not much additional service provided on top
	- But less networking stack software overheads as a result
	- Standard example: **User Datagram Protocol (UDP)**
- Connection-oriented transport layer
	- Provides error-free, reliable communication
	- Like having a UNIX pipe between processes on two different machines
	- Standard example: **Transmission Control Protocol (TCP)**

UDP – Connectionless service

- User Datagram Protocol
	- Essentially allows applications to send IP datagrams
	- With just slightly more encapsulation
- UDP transmits **segments**
	- Simply 8 byte header followed by payload

Ports

- Allows application-level multiplexing of network services
- Processes attach to ports to use network services
	- Port attachment is done with "BIND" operation
- Destination port
	- When a UDP packet arrives, its payload is handed to process attached to the destination port specified
- Source port
	- Mainly used when some reply is needed
	- Receiver can use the source port as the dest port in reply msg

UDP – What it does NOT do

- NO Flow control
- NO Error control
- NO Retransmission on receipt of bad segment
- User processes must handle this
- For apps needing precise control over packet flow, error control, or timing, UDP is a great fit
	- E.g. client-server situations where client sends short request and expects short reply back; client can timeout & retry easily
	- DNS (Domain Name System): For looking up IP addr of host name
		- Client sends host name, receives IP address response

TCP – Connection-oriented Service

- Transmission Control Protocol
	- Designed for end-to-end byte stream over unreliable network
	- Robust against failures and changing network properties
- TCP interface to user programs
	- Manages TCP streams and interfaces to the IP layer
	- Accepts user data streams from processes
	- Breaks up into pieces not larger than 64 KB
		- Often 1460 data bytes to fit in 1 Ethernet frame w/ IP + TCP headers
	- Sends each piece separately as IP datagram
	- Destination machine reconstructs original byte stream
	- Handles retransmissions & re-ordering
	- Provides error-free, reliable communication
- Result:
	- Can think of link like a pipe: Put data in one end, other side takes it out

TCP Service Model

- TCP service setup as follows:
	- Two endpoint processes create endpoints called sockets
	- Each socket has an address: IP address of host + 16-bit **port**
	- API functions used to create & communicate on sockets
- Ports
	- Numbers below 1024 called "well-known ports"
		- Reserved for standard services, like FTP, HTTP, SMTP <http://www.iana.org/assignments/service-names-port-numbers/service-names-port-numbers.xhtml>

TCP Service Model (2)

- TCP connections are full-duplex & point-to-point
	- Simultaneous traffic in both directions
	- Exactly 2 endpoints (no multicast or broadcast)
- TCP connection is a byte stream, not message stream
	- Receiver has no way to know what granularity bytes were sent
	- $-$ E.g. 4 x 512 byte writes vs. 1 x 2048 byte write
	- It can just receive some # of bytes at a time
	- Just like UNIX files!
- TCP may buffer data or send it immediately
	- PUSH flag indicates to TCP not to delay transmission
	- TCP tries to make a latency vs. bandwidth tradeoff

TCP Protocol

- TCP sequence number underlies much of the protocol – Every byte sent has its own **32-bit** sequence number
- TCP exchanges data in segments
	- 20-byte fixed header (w/ optional part)
	- Followed by 0 or more data bytes
	- TCP can merge writes into one segment or split a write up
	- Segment size limitations:
		- Must fit (including header) inside 65,515 byte IP payload
		- Networks have a MTU (max transfer unit) – e.g. 1500 bytes for Ethernet payload size
- Uses a sliding window protocol (acks + timeout + seq #)
	- Ack indicates the next seq # the receiver expects to get
	- May be piggy-backed with data going in the other direction

We're skipping this sort of stuff. If this were a networking course, we'd spend a looooong time on how TCP works.

TCP Header

We're skipping this sort of stuff. If this were a networking course, we'd spend a looooong time on how TCP works.

Layers 5+

- Communication of whatever you want
- Can use whatever transport(s) is(are) convenient/appropriate
- Freely structured
- Examples:
	- Skype (UDP)
	- \blacksquare SMTP = email (TCP)
	- \blacksquare HTTP = web (TCP)
	- Online games (TCP and/or UDP)

Below: Manually speaking HTTP to request http://google.com/ using the 'netcat' (nc) tool.

Demo: Wireshark

• Can observe packets in transit with network sniffer, e.g. Wireshark

Network layer summary

One more thing... VLANs

- Everyone on the same **layer 2 network** is in one "broadcast domain"
	- Equals one **IP subnet** (e.g. 192.168.0.0/24 would be one network)
	- No IP routing to go point-to-point; a network switch delivers directly
- What if we WANT a layer 2 boundary?
	- Just buy a physical switch for each network!

But we don't want to buy a different physical switch for each one!!

One more thing... VLANs

- *Logically* separate layer 2 networks
- Switch ports can be:
	- **Access ports:** can only see one VLAN, aren't aware of VLAN concept
	- **Trunk ports:** end point includes a VLAN tag in packet header to indicate which VLAN it wants to talk to; interprets such headers on incoming packets

Dynamic Host Configuration Protocol (DHCP)

(It's just one slide)

Dynamic Host Configuration Protocol (DHCP)

- DHCP: Allow hosts to enter a network and ask "what IP should I use for myself?"
- How it works:
	- 1. Client sends an IP broadcast "DISCOVERY" request (destination 255.255.255.255 UDP port 67)
	- 2. DHCP server on network sends an "OFFER" with IP address and other config (gateway router, DNS servers, maybe other stuff)
		- *Note: multiple offers might be provided by multiple DHCP servers (but usually it's just one)*
	- 3. Client sends a broadcast REQUEST for one of the offers
	- 4. DHCP server sends ACKNOWLEDGE back
	- 5. Client now has an IP address and basic config info
- DHCP can also be used to start network-boot (PXE), commonly used for diskless clusters, OS auto-install, etc.

Domain Name System (DNS)

(Many slides)

Purpose of DNS

- Map an easy-to-remember name to an IP address
- **Implications**
	- Without DNS, to send IP packet, must remember IP addresses manually! ...and they could change!
	- With DNS, we can use the name directly:
		- www.google.com or www.cnn.com
- DNS also provides inverse look-up that maps IP address to name

Before there was DNS...

- There was the HOSTS.TXT file (on Linux today as /etc/hosts)
- Maintained at SRI Network Information Center (NIC)
- Before DNS (1985), the name-to-IP address was done by downloading this single file from a central server with FTP
	- \blacksquare No hierarchical structure to the file
	- Still works on most OSes; can be used to define local names

Security implications... (Do the hosts + netcat demo)

Domain Namespace

- Domain namespace is a hierarchical and logical tree structure
- Label from a node to root in the DNS tree represents a DNS name
- Each subtree below a node is a DNS domain
	- DNS domain can contain hosts or other domains (subdomains)
- Examples of DNS domains: .edu, duke.edu, ece.duke.edu

Domain Namespace

Top-level domains

- Red is managed by Duke
- Green is managed by ECE
- Below top-level domain, administration of name space is delegated to organizations
	- Each organization can further delegate

Fully Qualified Domain Names

Top-level domains

- Every node in the DNS domain tree can be identified by a FQDN
	- Fully Qualified Domain Name
- FDQN (from right to left) consists of labels ("ece", "duke", "edu") separated by a period from the root to the node
- Each label can be up to 63 characters; full DNS name <= 255 chars
- FDQN contains characters, digits, dashes; not case-sensitive

Top-Level Domains

- Three types of top-level domains:
	- Generic Top Level Domains (gTLD)
		- 3 char code indicates the function of the organization
		- Use primarily within the US (e.g. gov, mil, edu, org, com, net)
	- Country Code Top Level Domain (ccTLD)
		- 2 char country or region code (e.g. us, jp, uk)
	- Reverse Domain
		- Special domain used for IP address-to-name mapping
		- in-addr.arpa
- More than 200 top-level domains

DNS Architecture

- Domain name space
	- Domain namespace is a hierarchical tree structure, a domain can be delegated to an organization
- Name servers
	- **Domain name hierarchy exists only in the abstract**
	- A host's name servers are specified in /etc/resolv.conf
- Resource records: (Name, Value, Type, Class, TTL)
	- Name ("duke.edu") and Value ("152.3.72.104")
	- Type specifies how the "Value" should be interpreted. Examples:
		- "NS" = Name Server: name is a domain and value is name of authoritative name server for this domain
		- "A" = Address: machine name and IP address
	- Class: record type (usually "IN" for Internet)
	- TTL: how long should the record be cached

Resource Records

Hierarchy of Name Servers

- Resolution of the hierarchical namespace is done by hierarchy of name servers
- Namespace is partitioned into zones. A zone is a contiguous portion of the DNS namespace
- Each server is responsible (authoritative for a zone)
- DNS server answers queries about host names in its zone

Name Servers

- Each zone has a primary and secondary name server
	- For reliability
	- Primary server maintains a zone file with zone info
		- Updates made to the primary server
	- Secondary server copies data stored at the primary server
- Adding a new host:
	- When new host is added (e.g. "newmachine.ece.duke.edu")
	- Administrator adds the IP info on the host (IP address, name) to a configuration file on the primary server

Root Name Servers

Map of the Root Servers

- Root name servers know how to find authoritative name servers for all top-level zones
- There are 13 (virtual) root name servers
- Root servers are critical for proper functioning of name resolution

Domain Name Resolution

- 1. User program issues a request for the IP address of a hostname gethostbyname()
- 2. Local resolver formulates a DNS query to the name server of the host
- 3. Name server checks if it is authorized to answer the query.
	- If yes, it responds.
	- Otherwise, it will query other name servers, starting at the root tree
- 4. When the name server has the answer it sends it to the resolver.

Inverse Query

- What is the host name for IP address 150.45.38.128?
	- IP address is converted to domain name: 128.38.45.150.in-addr.arpa
	- Resolver sends query for this address

Caching

- To reduce DNS traffic, name servers cache info
	- e.g. Domain name / IP address mappings
- When entry for a query is cached, the server does not contact other servers
- Note: if an entry is sent from a cache, the reply from the server is marked as "unauthoritative"
- Caching-only servers
	- Only purpose is to cache results
	- Do not contain zone info or a zone database file
- Negative responses will be cached too:
	- Caching "name in question does not exist"
	- Caching "Name in record exists, but requested data does not"

Modern follow-ons

- DNS with DHCP integration
	- When a new host uses DHCP to get on the network, the DHCP server can tell the DNS server about it, then the DNS server can answer requests for that host by name
- **Multicast DNS (mDNS)** and **Link-Local Multicast Name Resolution (LLMNR)**
	- Resolve hostnames when there's no local DNS server
	- Allows "automagic" host discovery on individual networks
	- **E** Zero configuration $-$ they're self-organizing protocols
- DNSSEC: DNS security (next slides)
	- Provides integrity, not confidentiality
- DNS over HTTPS (DoH): Very new standard, also provides encryption
	- Provides integrity *and* confidentiality

DNSSEC

- Problem: how do I know this DNS record I got is authentic?
- **DNSSEC**: A commonly deployed protocol to provide DNS integrity
	- "Sign" the record with asymmetric cryptography, use a "chain of trust" to show that the signature is valid.
	- \blacksquare (We haven't covered these concepts yet we'll see it when we cover crypto)
- DNSSEC *doesn't* encrypt requests (eavesdropping attacker can still see requests)

DNS security issues covered by DNSSEC

Quick intro to socket programming

Sockets

- How do user programs request to interact with networks?
- We can program using **network sockets**
	- For creating connections and sending / receiving messages
	- Often follows a client / server pattern

Client-Server Model

- Common communication model in networked systems
	- Client typically communicates with a server
	- Server may connect to multiple clients at a time
- Client needs to know:
	- Existence of a server providing the desired service
	- Address (commonly IP address) of the server
- Server does not need to know either about the client

Client-Server Overview

• Client and Server communicating across Ethernet using TCP/IP

Networking concept reminders

- Network interface is identified by an IP address
	- Or a hostname, which translates into an IP address
	- E.g. 127.0.0.1, localhost or login.oit.duke.edu
- Interface has 65536 ports (0-65535)
- Processes attach to ports to use network services
	- Port attachment is done with **bind()** operation
- Allows application-level multiplexing of network services
	- E.g. SSH vs. Web vs. Email may all use different ports
	- Many ports are standard (e.g. 80 for web server, 22 for SSH)
	- You may have seen URLs like http://127.0.0.1:4444
		- 127.0.0.1 is the IP, 4444 is the port

TCP Socket API

Example – UNIX TCP sockets

- Let's look at example code…
- Here is a great reference for use of socket-related calls
	- **<u><http://beej.us/guide/bgnet/></u>**

Server-Side Structure

• Often follows a common pattern to serve incoming requests

```
pid_t pid;
int listenfd, connfd;
listend = socket(...);/***fill the socket address with server's well known port***/
bind(listenfd, ...);
listen(listenfd, ...);
for ( ; ; ) {
  connfd = accept(listenfd, ...); // blocking callif ((pid = fork()) == 0) { // create a child process to service
     close(listenfd); // child closes listening socket
     /***process the request doing something using connfd ***/
     /* ................. */
     close(connfd);
    exit(0); // child terminates
 }
   close(connfd); // parent closes connected socket
}
```