

# Networking (2 parts)

- ♦ Goals:
  - \* Review networking basics
  - ♦ Discuss APIs
  - Trace how a packet gets from the network device to the application (and back)
  - \* Understand Receive livelock and NAPI

# 4 to 7 layer diagram (from Understanding Linux Network Internals) 7 Application 6 Presentation 5 Session 4 Transport 3 Network 3 Network 1 Physical 1/2 Hots-to-network (fithemet. . . ) 061 Figure 13-1. OSI and TCP/IP models

#### Nomenclature

- + Frame: hardware
- ♦ Packet: IP
- ♦ Segment: TCP/UDP
- \* Message: Application

# TCP/IP Reality

- \* The OSI model is great for undergrad courses
- \* TCP/IP (or UDP) is what the majority of programs use
  - ✦ Some random things (like networked disks) just use ethernet + some custom protocols

# Ethernet (or 802.2 or 802.3)

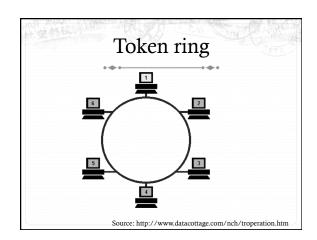
- \* All slight variations on a theme (3 different standards)
- \* Simple packet layout:
  - Header: Type, source MAC address, destination MAC address, length, (and a few other fields)
  - \* Data block (payload)
  - ♦ Checksum
- + Higher-level protocols "nested" inside payload
- + "Unreliable" no guarantee a packet will be delivered

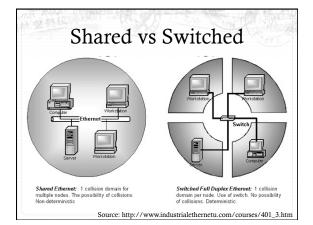
#### Ethernet History

- ♦ Originally designed for a shared wire (e.g., coax cable)
- \* Each device listens to all traffic
  - ♦ Hardware filters out traffic intended for other hosts
    - + I.e., different destination MAC address
  - \* Can be put in "promiscuous" mode, and record everything (called a network sniffer)
- Sending: Device hardware automatically detects if another device is sending at same time
  - \* Random back-off and retry

# Early competition

- + Token-ring network: Devices passed a "token" around
  - → Device with the token could send; all others listened
  - ♦ Like the "talking stick" in a kindergarten class
- Send latencies increased proportionally to the number of hosts on the network
  - Even if they weren't sending anything (still have to pass the token)
- Ethernet has better latency under low contention and better throughput under high





#### Switched networks

- \* Modern ethernets are switched
- ♦ What is a hub vs. a switch?
  - \* Both are a box that links multiple computers together
  - Hubs broadcast to all plugged-in computers (let computers filter traffic)
  - \* Switches track who is plugged in, only send to expected recipient
    - ♦ Makes sniffing harder ⊗

## Internet Protocol (IP)

- \* 2 flavors: Version 4 and 6
  - ♦ Version 4 widely used in practice---today's focus
- Provides a network-wide unique device address (IP address)
- This layer is responsible for routing data across multiple ethernet networks on the internet
  - Ethernet packet specifies its payload is IP
  - At each router, payload is copied into a new point-to-point ethernet frame and sent along

# Transmission Control Protocol (TCP)

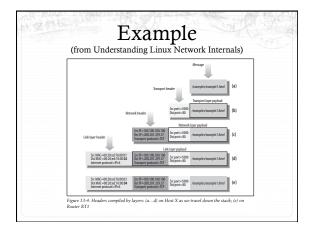
- Higher-level protocol that layers end-to-end reliability, transparent to applications
  - Lots of packet acknowledgement messages, sequence numbers, automatic retry, etc.
  - ♦ Pretty complicated
- \* Applications on a host are assigned a port number
  - \* A simple integer from 0-64k
  - \* Multiplexes many applications on one device
  - ♦ Ports below 1k reserved for privileged applications

# User Datagram Protocol (UDP)

- \* The simple alternative to TCP
  - \* None of the frills (no reliability guarantees)
- ♦ Same port abstraction (1-64k)
  - ♦ But different ports
  - ♦ I.e., TCP port 22 isn't the same port as UDP port 22

## Some well-known ports

- + 80 http
- ♦ 22 ssh
- ♦ 25 SMTP



## Networking APIs

- \* Programmers rarely create ethernet frames
- \* Most applications use the **socket** abstraction
  - $\begin{tabular}{ll} $\star$ & Stream of messages or bytes between two applications \\ \end{tabular}$
  - $\begin{tabular}{ll} $\bigstar$ & Applications still specify: protocol (TCP vs. UDP), remote host address \end{tabular}$ 
    - Whether reads should return a stream of bytes or distinct messages
- \* While many low-level details are abstracted, programmers must understand basics of low-level protocols

#### Sockets, cont.

- \* One application is the **server**, or **listens** on a predetermined port for new connections
- The client connects to the server to create a message channel
- The server accepts the connection, and they begin exchanging messages

#### Creation APIs

- int socket(domain, type, protocol) create a file handle representing the communication endpoint
  - \* Domain is usually AF\_INET (IP4), many other choices
  - \* Type can be STREAM, DGRAM, RAW
  - ♦ Protocol usually 0
- int bind(fd, addr, addrlen) bind this socket to a specific port, specified by addr
  - ♦ Can be INADDR\_ANY (don't care what port)

#### Server APIs

- int listen(fd, backlog) Indicate you want incoming connections
  - Backlog is how many pending connections to buffer until dropped
- int accept(fd, addr, len, flags) Blocks until you get a connection, returns where from in addr
  - \* Return value is a new file descriptor for child
  - → If you don't like it, just close the new fd

#### Client APIs

- \* Both client and server create endpoints using socket()
  - \* Server uses bind, listen, accept
  - \* Client uses connect(fd, addr, addrlen) to connect to server
- \* Once a connection is established:
  - ♦ Both use send/recv
  - ♦ Pretty self-explanatory calls

#### Client/server toy example

- + Quick demo ..
- ♦ Client/server code from

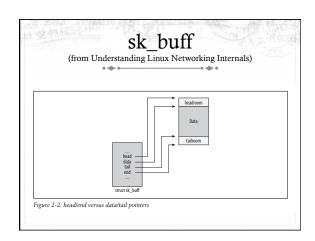
 $http://www.linuxhowtos.org/C\_C++/socket.htm\\$ 

# Linux implementation

- \* Sockets implemented in the kernel
  - \* So are TCP, UDP and IP
- ♦ Benefits:
  - Application doesn't need to be scheduled for TCP ACKs, retransmit, etc.
- \* Kernel trusted with correct delivery of packets
- \* A single system call (i386):
  - sys\_socketcall(call, args)
    - + Has a sub-table of calls, like bind, connect, etc.

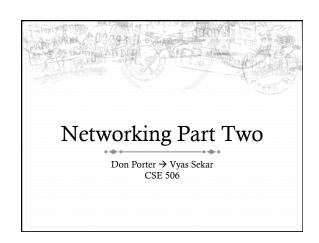
## Plumbing

- \* Each message is put in a sk\_buff structure
- Between socket/application and device, the sk\_buff is passed through a stack of protocol handlers
  - These handlers update internal bookkeeping, wrap payload in their headers, etc.
- At the bottom is the device itself, which sends/receives the packets



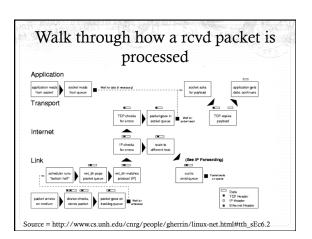
## Efficient packet processing

- \* Moving pointers is more efficient than removing headers
- \* Appending headers is more efficient than re-copy



# Recap from last class

- → Layering
- ‡ L2, L3, L4 basics
- \* Packet walkthrough



#### Interrupt handler

- + "Top half" responsible to:
  - \* Allocate a buffer (sk\_buff)
  - ♦ Copy received data into the buffer
  - ♦ Initialize a few fields
  - ♦ Call "bottom half" handler
- In some cases, sk\_buff can be pre-allocated, and network card can copy data in (DMA) before firing the interrupt
  - ♦ Lab 6 will follow this design

#### Quick review

- ♦ Why top and bottom halves?
  - To minimize time in an interrupt handler with other interrupts disabled
  - \* Gives kernel more scheduling flexibility
  - Simplifies service routines (defer complicated operations to a more general processing context)

## Digression: Softirqs

- \* A hardware IRQ is the hardware interrupt line
  - \* Also used for hardware "top half"
- \* Soft IRQ is the associated software "interrupt" handler
  - ♦ Or, "bottom half"
- How are these implemented in Linux?
  - \* Two canonical ways: Softirq and Tasklet
  - ♦ More general than just networking

#### Softirqs

- \* Kernel's view: per-CPU work lists
  - \* Tuples of <function, data>
- \* At the right time, call function(data)
  - \* Right time: Return from exceptions/interrupts/sys. calls
  - Also, each CPU has a kernel thread ksoftirqd\_CPU# that processes pending requests
  - \* ksoftirqd is nice +19. What does that mean?
    - ♦ Lowest priority only called when nothing else to do

#### Softirqs, cont.

- ♦ Device programmer's view:
  - Only one instance of a softirq function will run on a CPU at a time
    - + Doesn't need to be reentrant
      - reentrant if it can be interrupted in the middle of its execution and then safely called again ("re-entered") before its previous invocations complete execution
    - \* If interrupted, won't be called again by interrupt handler
      - + Subsequent calls enqueued!
  - \* One instance can run on each CPU concurrently, though
    - ♦ Must use locks

#### **Tasklets**

- \* For the faint of heart (and faint of locking prowess)
- \* Constrained to only run one at a time on any CPU
  - \* Useful for poorly synchronized device drivers
    - \* Say those that assume a single CPU in the 90's
  - Downside: If your driver uses tasklets, and you have multiple devices of the same type---the bottom halves of different devices execute serially

## Softirq priorities

- Actually, there are 6 queues per CPU; processed in priority order:
  - # HI\_SOFTIRQ (high/first)
  - → TIMER
  - ► NET TX
  - ♦ NET RX

  - → TASKLET (low/last)

## Observation 1

- → Devices can decide whether their bottom half is higher or lower priority than network traffic (HI or TASKLET)
  - \* Example: Video capture device may want to run its bottom half at HI, to ensure quality of service
  - \* Example: Printer may not care

#### Observation 2

- + Transmit traffic prioritized above receive. Why?
  - The ability to send packets may stem the tide of incoming packets
    - Obviously eliminates retransmit requests based on timeout
    - + Can also send "back-off" messages

#### Receive bottom half

- \* For each pending sk\_buff:
  - \* Pass a copy to any taps (sniffers)
  - \* Do any MAC-layer processing, like bridging
  - \* Pass a copy to the appropriate protocol handler (e.g., IP)
    - \* Recur on protocol handler until you get to a port
      - + Perform some handling transparently (filtering, ACK, retry)
    - + If good, deliver to associated socket
    - → If bad, drop

## Socket delivery

- Once the bottom half/protocol handler moves a payload into a socket:
  - Check and see if the task is blocked on input for this socket
  - ♦ If so, wake it up
- ♦ Read/recv system calls copy data into application

## Socket sending

- \* Send/write system calls copy data into socket
  - \* Allocate sk\_buff for data
  - \* Be sure to leave plenty of head and tail room!
- ♦ System call does protocol handling during application's timeslice
  - \* Note that receive handling done during ksoftirqd timeslice
- \* Last protocol handler enqueues a softirq to transmit

#### Transmission

- Softirq can go ahead and invoke low-level driver to do a send
- ♦ Interrupt usually signals completion
  - ♦ Interrupt handler just frees the sk\_buff

## Switching gears

- ♦ We've seen the path network data takes through the kernel in some detail
- Now, let's talk about how network drivers handle heavy loads

## Our cup runneth over

- \* Suppose an interrupt fires every time a packet comes in
  - \* This takes N ms to process the interrupt
- What happens when packets arrive at a frequency approaching or exceeding N?
- Will the bottom halves for any of these packets get executed?
  - ✦ No. They are lower-priority than new packets

## Receive livelock

- The condition that the system never makes progress because it spends all of its time starting to process new packets
- Real problem: Hard to prioritize other work over interrupts
- \* Principle: Better to process one packet to completion than to run just the top half on a million

# Receive livelock in practice | Mithout screend | Mithout screend

## Shedding load

- If you can't process all incoming packets, you must drop some
- Principle: If you are going to drop some packets, better do it early!
- If you quit taking packets off of the network card, the network card will drop packets once its buffers get full

#### Idea

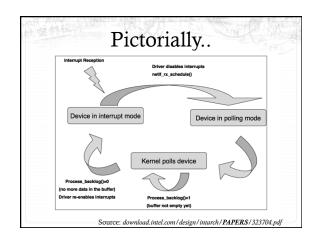
- ♦ Under heavy load, disable the network card's interrupts
- ♦ Use polling instead
  - \* Ask if there is more work once you've done the first batch
- This allows a packet to make it all the way through all of the bottom half processing, the application, and get a response back out
- \* Ensuring some progress! Yay!

## Why not poll all the time?

- \* If polling is so great, why even bother with interrupts?
- Latency: When incoming traffic is rare, we want highpriority, latency-sensitive applications to get their data ASAP

## General insight

- + If the expected input rate is low, interrupts are better
- \* When the expected input rate gets above a certain threshold, polling is better
- Just need to figure out a way to dynamically switch between the two methods...



# Why haven't we seen this before?

- + Why don't disks have this problem?
- ♦ Inherently rate limited
- If the CPU is bogged down processing previous disk requests, it can't issue more
- An external CPU can generate all sorts of network inputs

#### Linux NAPI

- ♦ Or New API. Seriously.
- Every driver provides a poll() method that does the lowlevel receive
  - \* Called in first step of softirq RX function
- ♦ Top half just schedules poll() to do the receive as softirq
  - Can disable the interrupt under heavy loads; use timer interrupt to schedule a poll
  - Bonus: Some rare NICs have a timer; can fire an interrupt periodically, only if something to say!

#### **NAPI**

- ♦ Gives kernel control to throttle network input
- ♦ Slow adoption means some measure of driver rewriting
- \* Backwards compatibility solution:
  - + Old top half still creates sk\_buffs and puts them in a queue
  - ♦ Queue assigned to a fake "backlog" device
  - ♦ Backlog poll device is scheduled by NAPI softirq
  - \* Interrupts can still be disabled

# NAPI Summary

- \* Too much input is a real problem
- NAPI lets kernel throttle interrupts until current packets processed
- ❖ Softirq priorities let some devices run their bottom halves before net TX/RX
  - \* Net TX handled before RX

## General summary

- \* Networking basics and APIs
- ♦ Idea of plumbing from socket to driver
  - \* Through protocol handlers and softirq poll methods
- \* NAPI and input throttling