Internetworking: philosophy, addressing, forwarding, resolution, fragmentation

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Based in part upon the slides of Prof. Raj Jain (OSU), J.Kurose (Umass), S. Keshav (Cornell), I.Stoica (UCB), S. Deering (Cisco)



Internetworking: heterogeneity & scale
 IP solution:

- Provide new packet format and overlay it on subnets.
- Ideas: Hierarchical address, address resolution, fragmentation/re-assembly, packet format design, forwarding algorithm etc
- Chapter 3,4,5,7 in Comer
- Reading: Clark: "The Design Philosophy of the DARPA Internet Protocols":
- Reading: Cerf, Kahn: "A Protocol for Packet Network Intercommunication"
- Reading: Mogul etal: "Fragmentation Considered Harmful"
- Reading: Addressing 101: Notes on Addressing: In PDF | In MS Word
- <u>Reading:</u> Notes for Protocol Design, E2e Principle, IP and Routing: <u>In PDF</u>
- Reference: RFC 791: Internet Protocol (IP) Spec.: In HTML Shivkumar Kalyanaraman

The Problem

 Before Internet: different packet-switching networks (e.g., ARPANET, ARPA packet radio)
 only nodes on the same network could communicate



A Translation-based Solution



application-layer gateways

- inevitable loss of some semantics
- difficult to deploy new internet-wide applications
- hard to diagnose and remedy end-to-end problems
- stateful gateways inhibited dynamic routing around failures

no global addressability

ad-hoc, application-specific solutions Shivkumar Kalyanaraman

The Internetworking Problem
 Two nodes communicating across a "*network of networks*"…

B

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How to transport packets through this heterogeneous mass ?

Declared Goal

"...both economic and technical considerations lead us to prefer that the interface be as simple and reliable as possible and deal primarily with passing data between networks using different packet switching strategies"

V. G. Cerf and R. E. Kahn, 1974

The Challenge: Heterogeneity

Share resources of different packet switching networks
interconnect existing networks

... but, packet switching networks differ widely
 different <u>services</u>

 e.g., degree of reliability

 different <u>interfaces</u>

 e.g., length of the packet that can be transmitted, address format
 different <u>protocols</u>

 e.g., routing protocols

The Challenge: Scale

Allow universal interconnection
 Mantra: Connectivity is its own reward

... but, core protocols had scalability issues
 Routing algorithms were limited in the number of nodes/links they could handle and were unstable after a point
 Universal addressing to go with routing
 As large numbers of users are multiplexed on a shared system, a congestion control paradigm is necessary for stability

□ No universal, scalable naming system...

The Internetworking Problem

Problems: <u>heterogeneity and scaling</u>
 <u>Heterogeneity</u>:

How to interconnect a large number of disparate *networks*? (lower layers)
 How to support a wide variety of *applications*?

(upper layers)

□ <u>Scaling:</u>

How to support a large number of end-nodes and applications in this interconnected network?

Solution



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The IP Solution ...



- internet-layer gateways & global addresses
- simple, application-independent, lowest denominator network service: best-effort datagrams
- stateless gateways could easily route around failures
- with application-specific knowledge out of gateways:
 - NSPs no longer had monopoly on new services
 - Internet: a platform for rapid, competitive innovation Shivkumar Kalyanaraman

Network-layer Overlay model

Define a *new* protocol (IP) and map all applications/networks to IP
 Require *only one mapping* (IP -> new protocol) when a new protocol/app is added
 Global address space can be created for universal addressibility and scaling

Before IP

(FTP – File Transfer Protocol, NFS – Network File Transfer, HTTP – World Wide Web protocol)



No network level overlay: each new application has to be re-implemented for every network technology!

IP

Key ideas: Overlay: better than any⇔any translation. Fewer, simpler mappings. Network-layer: efficient implementation, global addressing



What About the Future ?

Internet is running out of addresses
 Solutions

Classless Inter Domain Routing (CIDR)
 Network Address Translator (NATs)
 Dynamic Address Assignments

□IPv6

Why not variable-sized addresses?

Service to Apps

Unbounded but finite length messages
 byte streaming (What are the advantages?)
 Reliable and in-sequence delivery
 Full duplex

Solution: Transmission Control Protocol (TCP)

Original TCP/IP (Cerf & Kahn)

No separation between transport (TCP) and network (IP) layers

One common header

use ports to multiplex multiple TCP connections on the same host

32	32	16	16	<mark>8</mark> n
Source/Port	Source/Port	Window	ACK	Text

Byte-based sequence number (Why?)
 Flow control, but not congestion control

Today's TCP/IP

Separate transport (TCP) and network (IP) layer (why?)

split the common header in: TCP and UDP headers

fragmentation reassembly done by IP
 Congestion control (later in class)

IP Datagram Format



IP Datagram Format (Continued) First Word purpose: info, variable size header & packet. Version (4 bits) Internet header length (4 bits): <u>units of 32-bit</u> words. Min header is 5 words or 20 bytes. Type of service (TOS: 8 bits): Reliability, precedence, delay, and throughput. Not widely supported Total length (16 bits): header + data. Units of bytes. Total must be less than 64 kB.

IP Header (Continued)

2nd Word Purpose: fragmentation Identifier (16 bits): Helps uniquely identify the datagram between any source, destination address Flags (3 bits): More Flag (MF):more fragments Don't Fragment (DF) Reserved Fragment offset (13 bits): In units of 8 bytes

ar ragment onset (15 bits). <u>In units of 6 bytes</u>

IP Header (Continued)

Third word purpose: demuxing, error/looping control, timeout.

Time to live (8 bits): Specified in router hops
 Protocol (8 bits): Next level protocol to receive the data: for de-multiplexing.

Header checksum (16 bits): 1's complement sum of all 16-bit words in the header.

Change header => modify checksum using 1's complement arithmetic.

Recall: Signed Representations

Sign Magnitude	One's Complement	Two's Complement
000 = +0	000 = +0	000 = +0
001 = +1	001 = +1	001 = +1
010 = +2	010 = +2	010 = +2
011 = +3	011 = +3	011 = +3
100 = -0	100 = - 3	100 = - 4
101 = - 1	101 = - 2	101 = - 3
110 = - 2	110 = - 1	110 = - 2
111 = - 3	111 = - 0	111 = - 1

One's complement addition: normal addition & increment of the total if there was a carry. Eg: 110 (i.e. -1) + 111 (i.e. 0) = 101 +1 = 110 (i.e. -1)

Header Format (Continued)

Source Address (32 bits): Original source. Does not change along the path Destination. Address (32 bits): Final destination. Does not change along the path. Options (variable length): Security, source route, record route, stream id (used for voice) for reserved resources, timestamp recording Padding (variable length): Makes header length a multiple of 4 Payload Data (variable length): Data + header <</p> 65,535 bytes

TCP Header



Sequence number, acknowledgement, and advertised window – used by sliding-window based flow control
 Flags (selected):

SYN, FIN – establishing/terminating a TCP connection

ACK – set when Acknowledgement field is valid

RESET – abort connection

TCP Header (Cont)

- Checksum 1's complement and is computed over TCP header
 - TCP data
 - Pseudo-header (from IP header)
 - Note: breaks the layering!

Source	address
Destinatio	on address
0 Protocol (TCP)	TCP Segment length

TCP Connection Establishment

Three-way handshake
 Goal: agree on a set of parameters: the start sequence number for each side



IP Forwarding (I)

Source & Destination in same network (direct connectivity)

Recognize that destination IP address is on same network.^[1]

Find the destination LAN address. ^[2]

- Send IP packet encapsulated in LAN frame directly to the destination LAN address.
 - Encapsulation => source/destination IP addresses don't change

IP Forwarding (II)

B) Source & Destination in different networks (indirect connectivity)

- Recognize that destination IP address is not on same network. ^[1]
- Look up destination IP address in a (L3 forwarding) table to find a match, called the next hop router IP address.
- Send packet encapsulated in a LAN frame to the LAN address corresponding to the IP address of the next-hop router. ^[2]

[1] Addressing

[1] How to find if destination is in the same network ?

 IP address = network ID + host ID.
 Source and destination network IDs match => same network (I.e. direct connectivity)
 Splitting address into multiple parts is called hierarchical addressing



[2] Address Resolution

[2]: How to find the LAN address corresponding to an IP address ?
 Address Resolution Problem.
 Solution: ARP, RARP (later in this slide set)

IP Forwarding: Example Scenario

routing table in A

IP datagram:

misc	source	dest	
fields	IP addr	IP addr	data

datagram remains unchanged, as it travels source to destination addr fields of interest here



IP Forwarding (Direct)

misc	000444	000440	data
fields	223.1.1.1	223.1.1.3	uala

Starting at A, given IP datagram addressed to B:

look up net. address of B find B is on same net. as A link layer will send datagram directly to B inside link-layer frame

> B and A are <u>directly</u> <u>connected</u>



IP Forwarding (Indirect): Step 1

misc	000444	000400	data
fields	223.1.1.1	223.1.2.2	aata

Starting at A, dest. E:

look up network address of E E on *different* network

A, E not directly

attached

routing table: next hop router to E is 223.1.1.4 link layer sends datagram to

router 223.1.1.4 inside link-layer frame

datagram arrives at 223.1.1.4 continued.....



IP Forwarding (Indirect): Step 2

misc	000444	000 4 0 0	data
fields	223.1.1.1	223.1.2.2	data

Arriving at 223.1.4, destined for 223.1.2.2

look up network address of E E on *same* network as router's interface 223.1.2.9

router, E directly

attached

link layer sends datagram to 223.1.2.2 inside link-layer frame via interface 223.1.2.9 datagram arrives at 223.1.2.2

Dest	t.	next			
netwo	ork	route	^r Nhop	s ir	Iterface
223.1.	1	-	1	223.	1.1.4
223.1.2	2	-	1	223.	.1.2.9
223.1.	3	-	_ 1	223	.1.3.27
223.1.3 223.1.1.1 223.1.1.2 223.1.1.2 223.1.1.4 223.1.1.3 223.1.3 223.1.3.1		223 223.1.2 3.27 22 3.27 22	3.1.2.7 2.9 23.1.2 3.1.3.	2.2	

The Internet Network layer

Host, router network layer functions:


IP Addressing: introduction

- IP address: 32-bit identifier for host, router *interface Interface:* connection between host, router and physical link
 - router's typically have multiple interfaces
 - host may have multiple interfaces
 - IP addresses associated with interface, not host, router

Hosts in the same network have same network ID



IP Address Formats



Dotted Decimal Notation

 Binary: 11000000 00000101 00110000 00000011
 Hex Colon: C0:05:30:03
 Dotted Decimal: 192.5.48.3

ClassRangeA0 through 127B128 through 191C192 through 223D224 through 239E240 through 255

Subnet Addressing

Classful addressing inefficient: Everyone wants class B addresses

Can we split class A, B addresses spaces and accommodate more networks ?

Need another level of hierarchy. Defined by "<u>subnet mask</u>", which in general specifies the sets of bits belonging to the network address and host address respectively

Host

Boundary is flexible, and defined by subnet mask

Network





Inter-domain Routing Without CIDR

204.71.0.0 204.71.1.0 204.71.2.0

204.71.255.0

Service

Provider

204.71.0.0 204.71.1.0 204.71.2.0

.

204.71.255.0

Global Internet Routing Mesh

Inter-domain Routing With CIDR

204.71.0.0 204.71.2.0 204.71.2.0 204.71.2.55.0 Service Provider 204.71.0.0/16 Service Provider 204.71.0.0/16 Shivkumar Kalyanaraman

Implication on Forwarding: Subnet Route table lookup: IF ((Mask[i] & Destination Addr) = = Destination[i]) Forward to NextHop[i] In theory, subnet mask can end on any bit. In practice, mask must have contiguous 1s followed by contiguous zeros. Routers do not support other types of masks. So, (Address, Mask) = (12.4.0.0, 255.254.0.0) may be written as 12.4.0.0/15

Route Table Lookup: Subnet Example



DestinationMaskNext Hop30.0.0.0255.0.0.040.0.0.740.0.0.0255.0.0.0Deliver direct128.1.0.0255.255.0.0Deliver direct192.4.10.0255.255.255.0128.1.0.9

Implication on Forwarding: Supernetting (CIDR)

Longest Prefix Match (Classless) Forwarding

Destination = 12.5.9.16 payload **Prefix Next Hop** Interface 10.14.11.33 ATM 5/0/9 OK 0.0.0.0/0 better 12.0.0/8 10.14.22.19 **ATM 5/0/8** Ethernet 0/1/3 12.4.0.0/15 10.1.3.77 even better best! 12.5.8.0/23 attached **Serial 1/0/7** IP Forwarding Table umar Kalyanaraman

Variable Length Subnet Mask (VLSM)

Basic subneting: refers to a fixed mask in addition to natural mask (i.e. class A, B etc).

I.e. only a single mask (eg:: 255.255.255.0) can be used for all networks covered by the natural mask.

VLSM: Multiple different masks possible in a single class address space.

Eg: 255.255.255.0 and 255.255.254.0 could be used to subnet a single class B address space.

Allows more efficient use of address space.

Example: Address Block: 128.20.224.0/20. Networks: 2 of size 1000 nodes each; 2 of size 500 nodes each; 3 of size 250 nodes each. 4 of size 50 nodes each. What are the allocations?

1000 nodes need 10 bits => 32 -10 =22 bit prefixes needed 128.20.1110 00 00. 0000 0000/22 = 128.20.224.0/22 128.20.1110 01 00. 0000 0000/22 = 128.20.228.0/22

500 nodes need 9 bits => 32 -9 =23 bit prefixes needed 128.20.1110100 0. 0000 0000/23 = 128.20.232.0/23 128.20.1110101 0. 0000 0000/23 = 128.20.234.0/23

250 nodes need 8 bits => 32 -8 =24 bit prefixes needed 128.20.11101100. 0000 0000/24 = 128.20.236.0/24 128.20.11101101. 0000 0000/24 = 128.20.237.0/24 128.20.11101110. 0000 0000/24 = 128.20.238.0/24 50 nodes need 6 bits => 32 -6 =26 bit prefixes needed Shivkumar Kalyanaraman

Addressing Summary

Unique IP address per interface

- Classful (A,B,C) => address allocation not efficient
- Hierarchical => smaller routing tables
- Provision for broadcast, multicast, loopback addresses
- Subnet masks allow "subnets" within a "network" => improved address allocation efficiency
- Supernet (CIDR) allows variable sized network ID allocation
- VLSM allows further efficiency

Forwarding Summary

Forwarding:

Simple "next-hop" forwarding.
Last hop forwards directly to destination
Best-effort delivery : No error reporting. Delay, out-of-order, corruption, and loss possible => problem of higher layers!
Forwarding vs routing: tables setup by separate algorithm (s)

What IP does NOT provide

- End-to-end data reliability & flow control (done by TCP or application layer protocols)
- Sequencing of packets (like TCP)
- Error detection in payload (TCP, UDP or other transport layers)
- Error reporting (ICMP)
- Setting up route tables (RIP, OSPF, BGP etc)
- Connection setup (it is connectionless)
- Address/Name resolution (ARP, RARP, DNS)
- Configuration (BOOTP, DHCP)
- Multicast (IGMP, MBONE)

Maximum Transmission Unit Each subnet has a maximum frame size Ethernet: 1518 bytes FDDI: 4500 bytes Token Ring: 2 to 4 kB Transmission Unit = IP datagram (data + header) Each subnet has a maximum IP datagram length (header + payload) = MTU

Net 1

U=1500

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Net 2

J=1000

Fragmentation

 Datagrams larger than MTU are fragmented
 Original header is copied to each fragment and then modified (fragment flag, fragment offset, length,...)

Some option fields are copied (see RFC 791)



Fragmentation Example





IHL = 5, ID = 111, More = 0 Offset = 0W, Len = 472B







Fragmentation Example (Continued)

Payload size 452 bytes needs to be transmitted
 across a Ethernet (MTU=1500B) and a SLIP line (MTU=280B)

- Length = 472B, Header = 20B => Payload = 452B
- Fragments need to be multiple of 8-bytes.
 Nearest multiple to 260 (280 -20B) is 256B
 First fragment length = 256B + 20B = 276B.
 Second fragment length = (452B- 256B) + 20B = 216B

Reassembly

Reassembly only at the final destination
Partial datagrams are discarded after a timeout
Fragments can be further fragmented along the path. Subfragments have a format similar to fragments.

Net 2

R2

 \Box Minimum MTU along a path \Rightarrow Path MTU

R1

S

Net 1

J=1500



D

Net 3

=1500

Further notes on Fragmentation

Performance: single fragment lost => entire packet useless. Waste of resources all along the way. Ref: Kent & Mogul, 1987 Don't Fragment (DF) bit set => datagram discarded if need to fragment. ICMP message generated: may specify MTU (default = 0) Used to determine Path MTU (in TCP & UDP) The transport and application layer headers do not appear in all fragments. Problem if you need to peep into those headers.

Resolution Problems and Solutions
 Indirection through addressing/naming => requires address/name resolution
 Problem is to map destination layer N address to its layer N-1 address to allow packet transmission in layer N-1.



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Resolution Problems and Solutions (Continued)

- 1. Direct mapping: Make the physical addresses equal to the host ID part.
 - Mapping is easy.
 - Only possible if admin has power to choose both IP and physical address.
 - Ethernet addresses come preassigned (so do part of IP addresses!).
 - Ethernet addresses are 48 bits vs IP addresses which are 32-bits.

ARP techniques (Continued)
2: Table Lookup: Searching or indexing to get MAC addresses
Similar to lookup in /etc/hosts for names
Problem: change Ethernet card => change table

IP AddressMAC Address197.15.3.10A:4B:00:00:07:08197.15.3.20B:4B:00:00:07:00197.15.3.30A:5B:00:01:01:03

ARP techniques (Continued) 3. Dynamic Binding: ARP The host broadcasts a request: "What is the MAC address of 127.123.115.08?" The host whose IP address is 127,123,115,08 replies back: "The MAC address for 127.123.115.08 is 8A-5F-3C-23-45-56₁₆" ARP responses cached; LRU + Entry Timeout

All three methods are allowed in TCP/IP networks.

ARP Message Format



Target Protocol Address (4 bytes)

Type: ARP handles many layer 3 and layer 2s
 Protocol Address type: 0x0800 = IP
 Operation: 1= Request, 2=Response
 ARP messages are sent directly to MAC layer

Back to Goals (Clark'88)

- Connect existing networks
 - initially ARPANET and ARPA packet radio network
- Survivability
 - ensure communication service even in the presence of network and router failures
- 2. Support multiple types of services
- 3. Must accommodate a variety of networks
- Allow distributed management
- 5. Allow host attachment with a low level of effort
- 6. Be cost effective
- 7. Allow resource accountability

1. Survivability

- Continue to operate even in the presence of network failures (e.g., link and router failures)
 - as long as the network is not partitioned, two endpoint should be able to communicate...moreover, any other failure (excepting network partition) should be transparent to endpoints
- Decision: maintain state only at end-points (*fate-sharing*)
 - eliminate the problem of handling state inconsistency and performing state restoration when router fails

Internet: stateless network architecture

2. Types of Services

Add UDP to TCP to better support other types of applications

e.g., "real-time" applications This was arguably the main reasons for separating TCP and IP Provide datagram abstraction: lower common denominator on which other services can be built service differentiation was considered (remember ToS?), but this has never happened on the large scale (Why?)

3. Variety of Networks

Very successful (why?)

because the minimalist service; it requires from underlying network only to deliver a packet with a "reasonable" probability of success

...does not require:

reliability

in-order delivery

The mantra: IP over everything

Then: ARPANET, X.25, DARPA satellite network..

□ Now: ATM, SONET, WDM...

Other Goals

Allow distributed management

- Remember that IP interconnects networks
 - each network can be managed by a different organization
 - different organizations need to interact only at the boundaries
 - ... but this model complicates routing
- Cost effective
 - sources of inefficiency
 - header overhead
 - retransmissions
 - Routing
 - ...but routers relatively simple to implement (especially software side)
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Other Goals (Cont)

Low cost of attaching a new host \Box not a strong point \rightarrow higher than other architecture because the intelligence is in hosts (e.g., telephone vs. computer) bad implementations or malicious users can produce considerably harm (remember fatesharing?) Accountability very little so far

What About the Future?

Datagram not the best abstraction for:

resource management, accountability, QoS

A new abstraction: flow?
Routers require to maintain per-flow state (what is the main problem with this raised by Clark?)

state management

Proposed Solution

 soft-state: end-hosts responsible to maintain the state
 Problem: increase in control-traffic to maintain state, unless efficiently piggybacked

Summary: Internet Architecture

Packet-switched datagram network IP is the glue (network layer overlay) Hourglass architecture all hosts and routers run IP Stateless architecture no per flow state inside network



Summary: Minimalist Approach

Dumb network

- IP provide minimal functionalities to support connectivity
- addressing, forwarding, routing

Smart end system

- transport layer or application performs more sophisticated functionalities
- flow control, error control, congestion control

Advantages

- accommodate heterogeneous technologies (Ethernet, modem, satellite, wireless)
- support diverse applications (telnet, ftp, Web, X windows)
- decentralized network administration Shivkumar Kalyanaraman

Connect Existing Networks

Existing networks: ARPANET and ARPA packet radio
 Decision: packet switching

 Existing networks already were using this technology

Packet switching -> store and forward router architecture

Internet: a packet switched communication network consisting of different networks connected by store-andforward routers
Summary

Internetworking Problem

 IP header: supports connectionless delivery, variable length pkts/headers/options, fragmentation/reassembly,
Fragmentation/Reassembly, Path MTU discovery.
ARP, RARP: address mapping
Internet architectural principles