Cisco CCNA Study Guide

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Foreword:

This study guide is intended to provide those pursuing the CCNA certification with a framework of what concepts need to be studied. This is **not** a comprehensive document containing all the secrets of the CCNA, nor is it a "braindump" of questions and answers.

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I sincerely hope that this document provides some assistance and clarity in your studies.

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Part I

General Networking Concepts

Section 1 - Introduction to Networking -

What is a Network?

A **network** is defined as devices connected together to share information and services. The types of data/services that can be shared on a network is endless - documents, music, email, websites, databases, printers, faxes, telephony, videoconferencing, etc.

Protocols are "rules" that govern the method by which devices share data and services. Protocols are covered in great detail in subsequent sections.

Basic Network Types

Networks are generally broken down into two types:

LANs (**Local Area Networks**) - a high-speed network that covers a relatively small geographic area, usually contained within a single building or campus. A LAN is usually under the administrative control of a single entity/organization.

WANs (Wide Area Networks) – The book definition of a WAN is a network that spans large geographical locations, usually to interconnect multiple LANs.

A more **practical definition** describes a WAN as a network that traverses a public network or commercial carrier, using one of several **WAN technologies.** Thus, a WAN can be under the administrative control of several entities or organizations, and does not need to "span large geographical distances."

Note: Occasionally, books will define a third type of network known as a **MAN** (**Metropolitan Area Network**). A MAN is defined as a network that spans several LAN's across a city-wide geographic area. The term "MAN" is less prevalent than either LAN or WAN.

Network "Architectures"

A **host** refers to any device that is connected to your network. Some define a host as any device that has been assigned a **network address**.

A host can serve one or more functions:

- A host can *request* data (often referred to as a **client**)
- A host can *provide* data (often referred to as a **server**)
- A host can both request *and* provide data (often referred to as a **peer**)

Because of these varying functions, multiple network "architectures" have been developed, including:

- Peer-to-Peer networks
- Client/Server networks
- Mainframe/Terminal networks

When using a **peer-to-peer** architecture, all hosts on the network can both *request* and *provide* data and services. For example, configuring two Windows XP workstations to share files would be considered a peer-to-peer network.

Though peer-to-peer networks are simple to configure, there are several key disadvantages to this type of architecture. First, data is **spread across multiple devices**, making it difficult to manage and back-up that data. Second, security becomes problematic, as you must configure individual permissions and user accounts on each host.

When using a **client/server** architecture, hosts are assigned specific roles. **Clients** request data and services stored on **Servers**. Connecting Windows XP workstations to a Windows 2003 domain controller would be considered a client/server network.

While client/server environments tend to be more complex than peer-to-peer networks, there are several advantages. With data now **centrally located** on a server or servers, there is only one place to manage, back-up, and secure that data. This simplified management allows client/server networks to scale much larger than peer-to-peer. The key disadvantage of client/server architecture is that it introduces a **single point of failure.**

<u>Network "Architectures" (continued)</u>

When using a **mainframe/terminal** architecture, often referred to as a **thin-client environment**, a single device (the mainframe) stores all data and services for the network. This provides the same advantage as a client/server environment – centralized management and security of data.

Additionally, the mainframe performs all processing functions for the **dumb terminals** (or **thin-clients**) that connect to the mainframe. The thin clients perform *no processing whatsoever*, but serve only as input and output devices into the mainframe. Put more simply, the mainframe handles all the "thinking" for the thin-clients.

A typical hardware thin-client consists of a keyboard/mouse, a display, and an interface card into the network. Software thin-clients are also prevalent, and run on top of a client operating system (such as Windows XP or Linux).

Windows XP's remote desktop is an example of a thin-client application.

Section 2 - OSI Reference Model -

Network Reference Models

As computer network communication grew more prevalent, the need for a consistent standard for vendor hardware and software became apparent. Thus, the first development of a **network reference model** began in the 1970's, spearheaded by an international standards organization.

A network reference model serves as a blueprint, dictating how network communication should occur. Programmers and engineers design products that adhere to these models, allowing products from multiple manufacturers to interoperate.

Network models are organized into several **layers**, with each layer assigned a specific networking function. These functions are controlled by **protocols**, which govern end-to-end communication between devices.

Without the framework that network models provide, all network hardware and software would have been proprietary. Organizations would have been locked into a single vendor's equipment, and global networks like the Internet would have been impractical or even impossible.

The two most widely recognized network reference models are:

- The Open Systems Interconnection (OSI) model
- The **Department of Defense** (**DoD**) model

The OSI model was the first true network model, and consisted of seven layers. However, the OSI model has become deprecated over time, replaced with more practical models like the TCP/IP (or DoD) reference model.

Network models are not physical entities. For example, there is no *OSI device*. Devices and protocols *operate* at a specific layer of a model, depending on the function. Not every protocol fits perfectly within a specific layer, and some protocols spread across several layers.

OSI Reference Model

The **Open Systems Interconnection (OSI) model** was developed in the 1970's and formalized in 1983 by the **International Organization for Standardization (ISO)**. It was the first networking model, and provided the framework governing how information is sent across a network.

The OSI Model (ISO standard 7498) consists of seven layers, each corresponding to a particular network function:

7	Application
6	Presentation
5	Session
4	Transport
3	Network
2	Data-link
1	Physical

Various mnemonics have been devised to help people remember the order of the OSI model's layers:

7	Application	All	Away
6	Presentation	P eople	P izza
5	Session	Seem	Sausage
4	Transport	To	Throw
3	Network	Need	Not
2	Data-link	D ata	D o
1	Physical	Processing	Please

The ISO further developed an entire protocol suite based on the OSI model; however, this **OSI protocol suite** was never widely implemented. More common protocol suites can be difficult to fit within the OSI model's layers, and thus the model has been mostly deprecated.

A more practical model was developed by the Department of Defense (DoD), and became the basis for the TCP/IP protocol suite (and subsequently, the Internet). The DoD model is explained in detail later in this guide.

The OSI model is still used predominantly for educational purposes, as many protocols and devices are described by what *layer* they operate at.

The Upper Layers

The top three layers of the OSI model are often referred to as the **upper layers**. Thus, protocols that operate at these layers are usually called **upper-layer protocols**, and are generally implemented in *software*.

The function of the upper layers of the OSI model can be difficult to visualize. The upper layer protocols do not fit perfectly within each layer; and several protocols function at multiple layers.

The **Application layer (Layer 7)** provides the actual interface between the user application and the network. The user directly interacts with this layer.

Examples of application layer protocols include:

- FTP (via an FTP client)
- HTTP (via a web-browser)
- SMTP (via an email client)
- Telnet

The **Presentation layer (Layer 6)** controls the *formatting* of user data, whether it is text, video, sound, or an image. The presentation layer ensures that data from the *sending* device can be understood by the *receiving* device.

Additionally, the presentation layer is concerned with the **encryption** and **compression** of data.

Examples of presentation layer formats include:

- Text (RTF, ASCII, EBCDIC)
- Music (MIDI, MP3, WAV)
- Images (GIF, JPG, TIF, PICT)
- Movies (MPEG, AVI, MOV)

The **Session** layer (**Layer 5**) establishes, maintains, and ultimately terminates connections between devices. Sessions can be **full-duplex** (send and receive simultaneously), or **half-duplex** (send *or* receive, but not simultaneously).

The four layers below the upper layers are often referred to as the **lower** layers, and demonstrate the true benefit of learning the OSI model.

The Transport Layer

The **Transport layer (Layer 4)** is concerned with the reliable transfer of data, end-to-end. This layer ensures (or in some cases, does *not* ensure) that data arrives at its destination without corruption or data loss.

There are two types of transport layer communication:

- **Connection-oriented** parameters must be agreed upon by both parties before a connection is established.
- Connectionless no parameters are established before data is sent.

Parameters that are negotiated by connection-oriented protocols include:

- Flow Control (Windowing) dictating how much data can be sent between acknowledgements
- Congestion Control
- Error-Checking

The transport layer does not actually *send* data. Instead, it **segments** data into smaller pieces for transport. Each segment is assigned a sequence number, so that the receiving device can reassemble the data on arrival.

Examples of transport layer protocols include **Transmission Control Protocol** (**TCP**) and **User Datagram Protocol** (**UDP**). Both protocols are covered extensively in another guide.

Sequenced Packet Exchange (SPX) is the transport layer protocol in the IPX protocol suite.

The Network Layer

The **Network layer** (**Layer 3**) has two key responsibilities. First, this layer controls the **logical addressing** of devices. Logical addresses are organized as a hierarchy, and are not hard-coded on devices. Second, the network layer determines the **best path** to a particular destination network, and routes the data appropriately.

Examples of network layer protocols include **Internet Protocol (IP)** and **Internetwork Packet Exchange (IPX)**. IP version 4 (IPv4) and IP version 6 (IPv6) are covered in nauseating detail in separate guides.

The Data-Link Layer

The **Data-Link layer** (**Layer 2**) actually consists of two sub-layers:

- Logical Link Control (LLC) sub-layer
- Media Access Control (MAC) sub-layer

The LLC sub-layer serves as the intermediary between the physical link and all higher layer protocols. It ensures that protocols like IP can function regardless of what type of physical link is being used.

Additionally, the LLC sub-layer can use flow-control and error-checking, either in conjunction with a transport layer protocol (such as TCP), or *instead* of a transport layer protocol (such as UDP).

The MAC sub-layer controls access to the physical medium, serving as mediator if multiple devices are competing for the same physical link. Specific technologies have various methods of accomplishing this (for example: **Ethernet** uses *CSMA/CD*, **Token Ring** utilizes a *token*).

The data-link layer *packages* the higher-layer data into **frames**, so that the data can be put onto the physical wire. This packaging process is referred to as **framing** or **encapsulation**. The encapsulation type used is dependent on the underlying data-link/physical technology (such as Ethernet, Token Ring, FDDI, Frame-Relay, etc.)

Included in this frame is a source and destination **hardware** (or **physical**) address. Hardware addresses usually contain no hierarchy, and are often hard-coded on a device. Each device *must* have a unique hardware address on the network.

The Physical Layer

The **Physical layer** (**Layer 1**) controls the transferring of bits onto the physical wire. Devices such as network cards, hubs, and cabling are all considered physical layer equipment.

Physical-layer devices are covered extensively in other guides.

Explanation of Encapsulation

As data is passed from the user application down the virtual layers of the OSI model, each of the lower layers adds a **header** (and sometimes a **trailer**) containing protocol information specific to that layer. These headers are called **Protocol Data Units** (**PDUs**), and the process of adding these headers is called **encapsulation**.

For example, the Transport layer adds a header containing flow control and sequencing information (when using TCP). The Network layer header adds logical addressing information, and the Data-Link header contains physical addressing and other hardware specific information.

The PDU of each layer is identified with a different term:

Layer	PDU Name
Application	-
Presentation	-
Session	-
Transport	Segments
Network	Packets
Data-Link	Frames
Physical	Bits

Each layer *communicates* with the corresponding layer on the receiving device. For example, on the sending device, hardware addressing is placed in a Data-Link layer header. On the receiving device, that Data-Link layer header is processed and stripped away before it is sent up to the Network and other higher layers.

Specific devices are often identified by the OSI layer the device *operates* at; or, more specifically, what *header* or *PDU* the device processes. For example, **switches** are usually identified as Layer-2 devices, as switches process *hardware* (usually MAC) address information stored in the **Data-Link** header of a frame.

Similarly, **routers** are identified as Layer-3 devices, as routers look for *logical* (usually IP) addressing information in the **Network** header of a packet.

OSI Reference Model Example

The following illustrates the OSI model in more practical terms, using a web browser as an example:

- At the **Application layer**, a web browser serves as the user interface for accessing websites. Specifically, *HTTP* interfaces between the web browser and the web server.
- The format of the data being accessed is a **Presentation layer** function. Common data formats on the Internet include *HTML*, *XML*, *PHP*, *GIF*, *JPG*, etc. Additionally, any *encryption* or *compression* mechanisms used on a webpage are a function of this layer.
- The **Session layer** establishes the connection between the requesting computer and the web server. It determines whether the communication is *half-duplex* or *full-duplex*.
- The *TCP* protocol ensures the reliable delivery of data from the web server to the client. These are functions of the **Transport layer**.
- The *logical* (in this case, *IP*) addresses configured on the client and web server are a **Network Layer** function. Additionally, the routers that determine the best path from the client to the web server operate at this layer.
- IP addresses are translated to *hardware* addresses at the **Data-Link** layer.
- The actual cabling, network cards, hubs, and other devices that provide the physical connection between the client and the web server operate at the **Physical layer**.

Example Protocols

IP and the DoD Model

The **Internet Protocol** (**IP**) was developed by the Department of Defense (DoD) during the late 1970's. It was included in a group of protocols that became known as the **TCP/IP protocol suite**.

The DoD developed their own networking model to organize and define the TCP/IP protocol suite. This became known as the **DoD Model**, and consists of four layers:

	OSI Model		DoD Model
7	Application		
6	Presentation	4	Application
5	Session		
4	Transport	3	Host-to-Host
3	Network	2	Internet
2	Data-link	1	Network Access
1	Physical] 1	Network Access

The DoD model's streamlined approach proved more practical, as several protocols spread across multiple layers of the OSI Model.

The following chart diagrams where protocols fit in the DoD model:

Layer

Application	FTP, HTTP, SMTP
Host-to-Host	TCP, UDP
Internet	IP
Network Access	Ethernet

Section 3 - Ethernet Standards -

What is Ethernet?

Ethernet has become the standard technology used in LAN networking. Over time, the Ethernet standard has evolved to satisfy bandwidth requirements, resulting in various IEEE "categories" of Ethernet:

- 802.3 Ethernet (10 Mbps)
- 802.3u Fast Ethernet (100 Mbps)
- 802.3z or 802.3ab Gigabit Ethernet (1000 Mbps)

Various subsets of these Ethernet categories exist, operating at various speeds, distances, and cable types:

Standard	Cable Type	Speed	Max. Distance
10base2	Coaxial (thinnet)	10 Mbps	185 meters
10base5	Coaxial (thicknet)	10 Mbps	500 meters
10baseT	Twisted-pair	10 Mbps	100 meters
100baseT	Twisted-pair	100 Mbps	100 meters
1000baseSX	Fiber (multi-mode)	1 Gbps	>500 meters
1000baseLX	Fiber (single-mode)	1 Gbps	> 3 km

Half-Duplex vs. Full-Duplex

Ethernet devices can operate either at **half-duplex**, or **full-duplex**. At half duplex, devices can either transmit or receive data, but not simultaneously. Full-duplex allows devices to both transmit and receive at the same time. Devices connected to a **hub** can only operate at **half-duplex**, whereas devices connected to a **switch** can operate at **full-duplex**.

Half-duplex Ethernet uses Carrier Sense Multiple Access with Collision Detect (CSMA/CD) to control media access. Devices monitor the physical link, and will only transmit a frame if the link is idle. If two devices send a packet simultaneously, a collision will occur. When a collision is detected, both NICs will wait a random amount of time before resending their respective packets. Full-duplex Ethernet does not use CSMA/CD.

Port speed and duplex can be either manually configured or auto-negotiated with a hub or switch. However, a **duplex mismatch** will occur if one side is configured manually, and the other configured for auto-negotiation.

Ethernet (10 Mbps)

The first incarnation of Ethernet operated at 10 Mbps, over thinnet (10base2), thicknet (10base5), or twisted pair (10baseT) mediums. Ethernet's specifications were outlined in the IEEE **802.3** standard.

Even though the term "Ethernet" is widely used to describe any form of Ethernet technology, technically the term refers to the 10 Mbps category. The most common implementation of Ethernet is over Category 5 twisted-pair cable, with a maximum distance of 100 meters.

Full Duplex Ethernet allows devices to both send and receive simultaneously, doubling the bandwidth to 20 Mbps per port. Only devices connected to a switch can operate at Full Duplex.

Fast Ethernet

Fast Ethernet, or **IEEE 802.3u**, operates at 100 Mbps, utilizing Category 5 twisted-pair (100base-TX) or fiber cabling (100base-FX).

Full Duplex Fast Ethernet allows devices connected to a switch to both send and receive simultaneously, doubling the bandwidth to 200 Mbps per port.

Many switches (and hubs) support both Ethernet and Fast Ethernet, and are commonly referred to as 10/100 switches. These switches will autonegotiate both port speed and duplex.

As mentioned earlier, it is also possible to statically configure this information. Both the device and switch must be configured for autonegotiation (or both configured with the same static settings), otherwise a **duplex mismatch** error will occur.

Gigabit Ethernet

Gigabit Ethernet operates at 1000 Mbps, and can be utilized over Category 5e twisted-pair (1000baseT) or fiber cabling (1000baseSX or 1000baseLX). Gigabit Ethernet over copper is defined in the **IEEE 802.3ab** standard.

Full Duplex Gigabit Ethernet allows devices connected to a switch to both send and receive simultaneously, doubling the bandwidth to 2000 Mbps.

Newer switches can support Ethernet, Fast Ethernet, and Gigabit Ethernet simultaneously, and are often referred to as 10/100/1000 switches. Again, switches and devices can auto-negotiate both speed and duplex.

10 Gigabit Ethernet has also been developed, defined in the **IEEE 802.3ae** standard, and currently can operate only over fiber cabling.

Twisted-Pair Cabling

Twisted-pair cable usually contains 2 or 4 pairs of wire, which are twisted around each other to reduce **crosstalk**. Crosstalk is a form of electromagnetic interference (EMI) or "noise" that reduces the strength and quality of a signal. It is caused when the signal from one wire "bleeds" or interferes with another wire's signal.

Twisted-pair cabling can be either shielded or unshielded. Shielded twisted-pair is more resistant to from external EMI. Florescent light ballasts, microwaves, and radio transmitters can all create EMI.

There are various categories of twisted-pair cable, identified by the number of "twists per inch."

- Category 3 (three twists per inch)
- Category 5 (five twists per inch)
- Category 5e (five twists per inch, pairs are twisted around each other)

Category 5 (and 5e) twisted-pair cabling usually contains four pairs of wire (eight wires total), and each wire is assigned a color:

- White Orange
- Orange
- White Green
- Green

- White Blue
- Blue
- White Brown
- Brown

Types of Twisted-Pair Cables

Various *types* of twisted-pair cables can be used. A **straight-through** cable is used in the following circumstances:

- From a host to a hub (or switch)
- From a router to a hub (or switch)

The pins (wires) on each end of a straight-through cable must be identical. The most common straight-through standard is wired as follows:

1	White Orange	 White Orange	1
2	Orange	 Orange	2
3	White Green	 White Green	3
4	Blue	 Blue	4
5	White Blue	 White Blue	5
6	Green	 Green	6
7	White Brown	 White Brown	7
8	Brown	 Brown	8

A **cross-over** cable is used in the following circumstances:

- From a host to a host
- From a hub to a hub
- From a switch to a switch
- From a hub to a switch
- From a router to a router

To make a crossover cable, we must swap pins 1 and 3, and pins 2 and 6 on one end of the cable. The most common crossover standard is as follows:

1	White Orange	 White Green	3
2	Orange	 Green	6
3	White Green	 White Orange	1
4	Blue	 Blue	4
5	White Blue	 White Blue	5
6	Green	 Orange	2
7	White Brown	 White Brown	7
8	Brown	 Brown	8

Remember, when connecting *anything* into a hub or switch, *except* for another hub or switch, we use a straight-through cable. When connecting "like" devices, we use a crossover cable.

* * *

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Types of Twisted-Pair Cables (continued)

Finally, a **roll-over** cable is used to connect a PC into a Cisco router's console or auxiliary port.

The pins are completely reversed on one end to make a rollover cable:

1	White Orange	 Brown	8
2	Orange	 White Brown	7
3	White Green	 Green	6
4	Blue	 White Blue	5
5	White Blue	 Blue	4
6	Green	 White Green	3
7	White Brown	 Orange	2
8	Brown	 White Orange	1

Section 4 - Hubs vs. Switches vs. Routers -

Layered Communication

Network communication models are generally organized into **layers.** The **OSI model** specifically consists of **seven layers**, with each layer representing a specific networking function. These functions are controlled by **protocols**, which govern end-to-end communication between devices.

As data is passed from the user application down the virtual layers of the OSI model, each of the lower layers adds a **header** (and sometimes a **trailer**) containing protocol information specific to that layer. These headers are called **Protocol Data Units** (**PDUs**), and the process of adding these headers is referred to as **encapsulation**.

The PDU of each lower layer is identified with a unique term:

#	Layer	PDU Name
7	Application	-
6	Presentation	-
5	Session	-
4	Transport	Segments
3	Network	Packets
2	Data-link	Frames
1	Physical	Bits

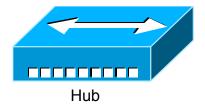
Commonly, network devices are identified by the OSI layer they *operate* at (or, more specifically, what *header* or *PDU* the device processes).

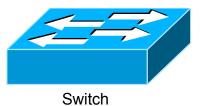
For example, **switches** are generally identified as Layer-2 devices, as switches process information stored in the **Data-Link** header of a frame (such as MAC addresses in Ethernet). Similarly, **routers** are identified as Layer-3 devices, as routers process *logical* addressing information in the **Network** header of a packet (such as IP addresses).

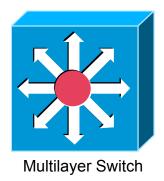
However, the strict definitions of the terms *switch* and *router* have blurred over time, which can result in confusion. For example, the term *switch* can now refer to devices that operate at layers higher than Layer-2. This will be explained in greater detail in this guide.

Icons for Network Devices

The following icons will be used to represent network devices for all guides on routeralley.com:









Layer-1 Hubs

Hubs are Layer-1 devices that physically connect network devices together for communication. Hubs can also be referred to as **repeaters**.

Hubs provide *no intelligent forwarding* whatsoever. Hubs are incapable of processing either Layer-2 or Layer-3 information, and thus cannot make decisions based on hardware or logical addressing.

Thus, hubs will always forward *every* frame out *every* port, excluding the port originating the frame. Hubs do not differentiate between frame types, and thus will always forward unicasts, multicasts, and broadcasts out *every* port but the originating port.

Ethernet hubs operate at **half-duplex**, which allows a device to either transmit or receive data, but not simultaneously. Ethernet utilizes **Carrier Sense Multiple Access with Collision Detect (CSMA/CD)** to control media access. Host devices monitor the physical link, and will only transmit a frame if the link is idle.

However, if two devices transmit a frame simultaneously, a **collision** will occur. If a collision is detected, the hub will discard the frames and signal the host devices. Both devices will wait a random amount of time before resending their respective frames.

Remember, if *any* two devices connected to a hub send a frame simultaneously, a collision *will* occur. Thus, all ports on a hub belong to the same **collision domain**. A collision domain is simply defined as any physical segment where a collision can occur.

Multiple hubs that are uplinked together still all belong to *one* collision domain. Increasing the number of host devices in a single collision domain will increase the number of collisions, which can significantly degrade performance.

Hubs also belong to only one **broadcast domain** – a hub will forward both broadcasts and multicasts out *every* port but the originating port. A broadcast domain is a logical segmentation of a network, dictating how far a broadcast (or multicast) frame can propagate.

Only a Layer-3 device, such as a router, can separate broadcast domains.

Layer-2 Switching

Layer-2 devices build **hardware address tables**, which will contain the following at a minimum:

- Hardware addresses for host devices
- The port each hardware address is associated with

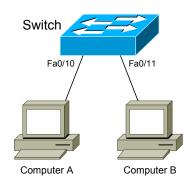
Using this information, Layer-2 devices will make intelligent forwarding decisions based on frame (Data-Link) headers. A frame can then be forwarded out *only* the appropriate destination port, instead of *all* ports.

Layer-2 forwarding was originally referred to as **bridging**. Bridging is a largely deprecated term (mostly for marketing purposes), and Layer-2 forwarding is now commonly referred to as **switching**.

There are some subtle technological differences between *bridging* and *switching*. Switches usually have a higher port-density, and can perform forwarding decisions at wire speed, due to specialized hardware circuits called **ASICs** (**Application-Specific Integrated Circuits**). Otherwise, bridges and switches are nearly identical in function.

Ethernet switches build **MAC-address tables** through a dynamic learning process. A switch behaves much like a hub when first powered on. The switch will flood every frame, including unicasts, out *every* port but the originating port.

The switch will then build the MAC-address table by examining the **source MAC address** of each frame. Consider the following diagram:



When ComputerA sends a frame to ComputerB, the switch will add *ComputerA*'s MAC address to its table, associating it with port fa0/10. However, the switch will not learn *ComputerB's* MAC address until ComputerB sends a frame to ComputerA, or to another device connected to the switch. Switches always learn from the source MAC address.

A switch is in a perpetual state of learning. However, as the MAC-address table becomes populated, the flooding of frames will decrease, allowing the switch to perform more efficient forwarding decisions.

<u>Layer-2 Switching (continued)</u>

While hubs were limited to half-duplex communication, switches can operate in **full duplex**. *Each individual port* on a switch belongs to its *own collision domain*. Thus, switches create **more collision domains**, which results in **fewer collisions**.

Like hubs though, switches belong to only *one* **broadcast domain.** A Layer-2 switch will forward both broadcasts and multicasts out *every* port but the originating port. Only Layer-3 devices separate broadcast domains.

Because of this, Layer-2 switches are poorly suited for large, scalable networks. The Layer-2 header provides no mechanism to differentiate one *network* from another, only one *host* from another.

This poses *significant* difficulties. If *only* hardware addressing existed, all devices would technically be on the *same* network. Modern internetworks like the Internet could not exist, as it would be impossible to separate *my* network from *your* network.

Imagine if the entire Internet existed purely as a Layer-2 switched environment. Switches, as a rule, will forward a broadcast out *every* port. Even with a conservative estimate of a billion devices on the Internet, the resulting broadcast storms would be devastating. The Internet would simply collapse.

Both hubs and switches are susceptible to **switching loops**, which result in destructive broadcast storms. Switches utilize the **Spanning Tree Protocol** (**STP**) to maintain a loop-free environment. STP is covered in great detail in another guide.

Remember, there are three things that switches do that hubs do not:

- Hardware address learning
- Intelligent forwarding of frames
- Loop avoidance

Hubs are almost entirely deprecated – there is no advantage to using a hub over a switch. At one time, switches were more expensive and introduced more latency (due to processing overhead) than hubs, but this is no longer the case.

Layer-2 Forwarding Methods

Switches support three **methods** of forwarding frames. Each method copies all or part of the frame into memory, providing different levels of latency and reliability. **Latency** is *delay* - less latency results in quicker forwarding.

The **Store-and-Forward** method copies the *entire* frame into memory, and performs a Cycle Redundancy Check (CRC) to completely ensure the integrity of the frame. However, this level of error-checking introduces the highest latency of any of the switching methods.

The **Cut-Through** (**Real Time**) method copies only enough of a frame's header to determine its destination address. This is generally the *first* 6 *bytes* following the preamble. This method allows frames to be transferred at *wire speed*, and has the least latency of any of the three methods. No error checking is attempted when using the cut-through method.

The **Fragment-Free** (**Modified Cut-Through**) method copies only the *first* 64 bytes of a frame for error-checking purposes. Most collisions or corruption occur in the first 64 bytes of a frame. Fragment-Free represents a compromise between reliability (store-and-forward) and speed (cut-through).

Layer-3 Routing

Layer-3 **routing** is the process of forwarding a packet from one *network* to another *network*, based on the Network-layer header. Routers build **routing tables** to perform forwarding decisions, which contain the following:

- The destination network and subnet mask
- The **next hop** router to get to the destination network
- Routing *metrics* and Administrative Distance

Note that Layer-3 forwarding is based on the destination *network*, and not the destination *host*. It is possible to have *host routes*, but this is less common.

The routing table is concerned with two types of Layer-3 protocols:

- **Routed protocols** assigns logical addressing to devices, and routes packets between networks. Examples include IP and IPX.
- **Routing protocols** dynamically builds the information in routing tables. Examples include RIP, EIGRP, and OSPF.

Each individual interface on a router belongs to its own collision domain. Thus, like switches, routers create **more collision domains**, which results in **fewer collisions**.

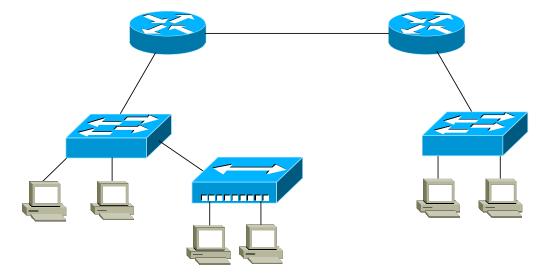
Unlike Layer-2 switches, Layer-3 routers also **separate broadcast domains**. As a rule, a router **will never forward broadcasts** from one network to another network (unless, of course, you explicitly configure it to). ©

Routers will not forward multicasts either, unless configured to participate in a multicast tree. Multicast is covered in great detail in another guide.

Traditionally, a router was required to copy each individual packet to its buffers, and perform a route-table lookup. Each packet consumed CPU cycles as it was forwarded by the router, resulting in latency. Thus, routing was generally considered **slower** than switching.

It is now possible for routers to *cache* network-layer flows in hardware, greatly reducing latency. This has blurred the line between *routing* and *switching*, from both a technological and marketing standpoint. Caching network flows is covered in greater detail shortly.

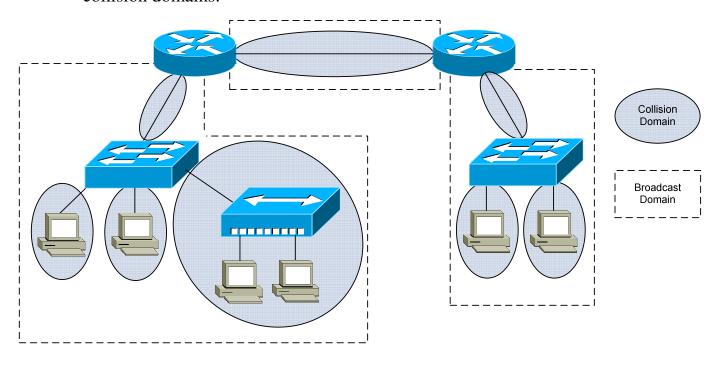
Collision vs. Broadcast Domain Example



Consider the above diagram. Remember that:

- Routers separate *broadcast* and *collision* domains.
- Switches separate *collision* domains.
- Hubs belong to only one *collision* domain.
- Switches and hubs both only belong to one *broadcast* domain.

In the above example, there are **THREE** broadcast domains, and **EIGHT** collision domains:



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VLANs - A Layer-2 or Layer-3 Function?

By default, a switch will forward both broadcasts and multicasts out *every* port but the originating port.

However, a switch can be logically segmented into multiple broadcast domains, using **Virtual LANs** (or **VLANs**). VLANs are covered in extensive detail in another guide.

Each VLAN represents a unique broadcast domain:

- Traffic between devices within the *same* VLAN is switched (forwarded at Layer-2).
- Traffic between devices in *different* VLANs requires a Layer-3 device to communicate.

Broadcasts from one VLAN will not be forwarded to another VLAN. This separation provided by VLANs is **not a Layer-3 function.** VLAN tags are inserted into the **Layer-2 header**.

Thus, a switch that supports VLANs is not necessarily a Layer-3 switch. However, a purely Layer-2 switch cannot route between VLANs.

Remember, though VLANs provide separation for *Layer-3* broadcast domains, and are often associated with IP subnets, they are still a *Layer-2* function.

Layer-3 Switching

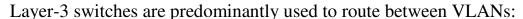
In addition to performing Layer-2 switching functions, a **Layer-3 switch** must also meet the following criteria:

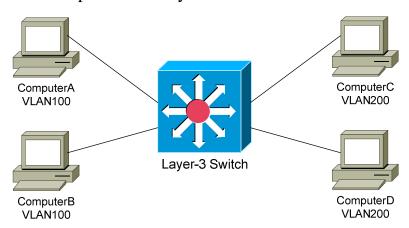
- The switch must be capable of making Layer-3 forwarding decisions (traditionally referred to as routing).
- The switch must cache network traffic flows, so that Layer-3 forwarding can occur in hardware.

Many older modular switches support Layer-3 route processors – this alone does not qualify as Layer-3 switching. Layer-2 and Layer-3 processors can act independently within a single switch chassis, with each packet requiring a route-table lookup on the route processor.

Layer-3 switches leverage ASICs to perform Layer-3 forwarding in hardware. For the first packet of a particular traffic flow, the Layer-3 switch will perform a standard route-table lookup. This flow is then *cached* in hardware – which preserves required routing information, such as the destination network and the MAC address of the corresponding next-hop.

Subsequent packets of that flow will bypass the route-table lookup, and will be forwarded based on the cached information, reducing latency. This concept is known as *route once*, *switch many*.





Traffic between devices within the same VLAN, such as ComputerA and ComputerB, is *switched* at Layer-2 as normal. The first packet between devices in different VLANs, such as ComputerA and ComputerD, is *routed*. The switch will then cache that IP traffic flow, and subsequent packets in that flow will be *switched* in hardware.

<u>Layer-3 Switching vs. Routing – End the Confusion!</u>

The evolution of network technologies has led to considerable confusion over the terms *switch* and *router*. Remember the following:

- The traditional definition of a *switch* is a device that performs Layer-2 forwarding decisions.
- The traditional definition of a *router* is a device that performs Layer-3 forwarding decisions.

Remember also that, switching functions were typically performed in *hardware*, and routing functions were typically performed in *software*. This resulted in a widespread perception that switching was *fast*, and routing was *slow* (and *expensive*).

Once Layer-3 forwarding became available in hardware, marketing gurus muddied the waters by distancing themselves from the term *router*. Though Layer-3 forwarding in hardware is still *routing* in every technical sense, such devices were rebranded as Layer-3 switches.

Ignore the marketing noise. A Layer-3 switch is still a router.

Compounding matters further, most devices still currently referred to as *routers* can perform Layer-3 forwarding in hardware as well. Thus, both Layer-3 switches *and* Layer-3 routers perform nearly identical functions at the same performance.

There are some differences in *implementation* between Layer-3 switches and routers, including (but not limited to):

- Layer-3 switches are optimized for Ethernet, and are predominantly used for inter-VLAN routing. Layer-3 switches can also provide Layer-2 functionality for intra-VLAN traffic.
- Switches generally have higher port densities than routers, and are considerably cheaper per port than routers (for Ethernet, at least).
- Routers support a large number of WAN technologies, while Layer-3 switches generally do not.
- Routers generally support more advanced feature sets.

Layer-3 switches are often deployed as the backbone of LAN or campus networks. Routers are predominantly used on network perimeters, connecting to WAN environments.

(Fantastic Reference: http://blog.ioshints.info/2011/02/how-did-we-ever-get-into-this-switching.html)

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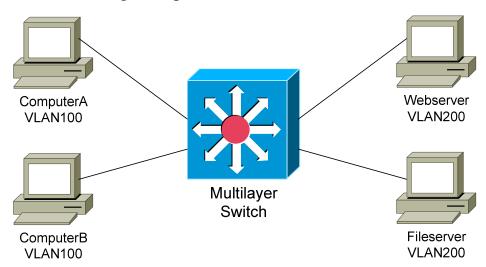
Multilayer Switching

Multilayer switching is a generic term, referring to any switch that forwards traffic at layers higher than Layer-2. Thus, a Layer-3 switch is considered a multilayer switch, as it forwards frames at Layer-2 and packets at Layer-3.

A **Layer-4 switch** provides the same functionality as a Layer-3 switch, but will additionally examine and cache **Transport-layer application flow** information, such as the TCP or UDP port.

By caching application flows, **QoS** (**Quality of Service**) functions can be applied to preferred applications.

Consider the following example:



Network and application traffic flows from ComputerA to the Webserver and Fileserver will be cached. If the traffic to the Webserver is preferred, then a higher QoS priority can be assigned to that application flow.

Some advanced multilayer switches can provide load balancing, content management, and other application-level services. These switches are sometimes referred to as **Layer-7 switches**.

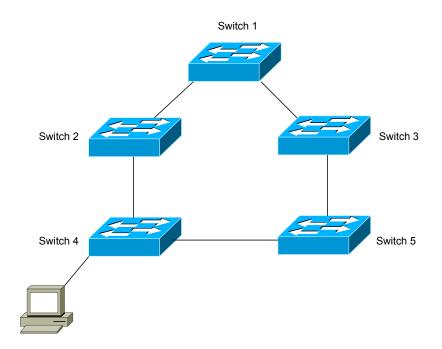
Section 5 - Spanning Tree Protocol -

Switching Loops

By default, a switch will forward a broadcast or multicast out all ports, excluding the port the broadcast/multicast was sent from.

When a *loop* is introduced into the network, a highly destructive **broadcast storm** can develop within seconds. Broadcast storms occur when broadcasts are endlessly switched through the loop, choking off all other traffic.

Consider the following looped environment:



If the computer connected to Switch 4 sends out a broadcast, the switch will forward the broadcast out all ports, including the ports connecting to Switch 2 and Switch 5. Those switches, likewise, will forward that broadcast out all ports, including to their neighboring switches.

The broadcast will loop around the switches infinitely. In fact, there will be *two* separate broadcast storms cycling in opposite directions through the switching loop. Only powering off the switch or physically removing the loop will stop the storm.

Spanning Tree Protocol (STP)

Switches (and bridges) needed a mechanism to prevent loops from forming, and thus **Spanning Tree Protocol (STP,** or **IEEE 802.1D)** was developed. STP is **enabled** by default on all VLANs on Catalyst switches.

STP-enabled switches communicate to form a topology of the entire switching network, and then shutting down (or *blocking*) a port if a loop exists. The blocked port can be reactivated if another link on the switching network goes down, thus preserving fault-tolerance. Once all switches agree on the topology database, the switches are considered **converged**.

STP switches send **BPDU's** (**Bridge Protocol Data Units**) to each other to form their topology databases. BPDU's are sent out all ports every **two seconds**, are forwarded to a specific MAC multicast address: 0180.c200.0000.

STP Types

Various flavors of 802.1D STP exist, including:

- Common Spanning Tree (CST) A single STP process is used for all VLANs.
- **Per-VLAN Spanning Tree (PVST)** Cisco proprietary version of STP, which employs a separate STP process for each VLAN.
- **Per-VLAN Spanning Tree Plus (PVST+)** Enhanced version of PVST that allows CST-enabled switches and PVST-enabled switches to interoperate. This is *default* on newer Catalyst switches.

The STP Process

To maintain a loop-free environment, STP performs the following functions:

- A **Root Bridge** is elected
- Root Ports are identified
- **Designated Ports** are identified
- If a loop exists, a port is placed in **Blocking** state. If the loop is removed the blocked port is activated again.

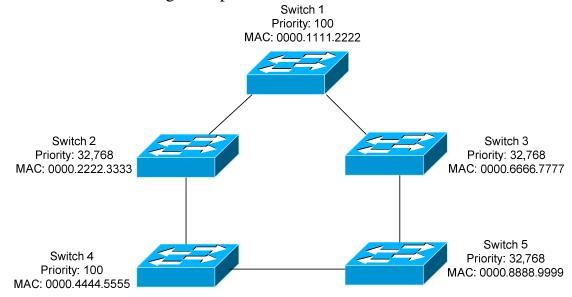
If multiple loops exist in the switching environment, multiple ports will be placed in a blocking state.

Electing an STP Root Bridge

The first step in the STP process is electing a **Root Bridge**, which serves as the centralized point of the STP topology. Good design practice dictates that the Root Bridge be placed closest to the center of the STP topology.

The Root Bridge is determined by a switch's **priority**. The default priority is **32,768**, and the **lowest priority wins**. In case of a tie in priority, the switch with the **lowest MAC address** will be elected root bridge. The combination of a switch's priority and MAC address make up that switch's **Bridge ID**.

Consider the following example:



Remember that the lowest priority determines the Root Bridge. Switches 2, 3, and 5 have the default priority set. Switches 1 and 4 each have a priority of 100 configured. However, Switch 1 will become the root bridge, as it has the lowest MAC address.

Switches exchange BPDU's to perform the election process. By default, all switches "believe" they are the Root Bridge, until a switch with a lower Bridge ID is discovered.

Root Bridge elections are a continuous process. If a new switch with a lower Bridge ID is added to the topology, it will be elected as the new Root Bridge.

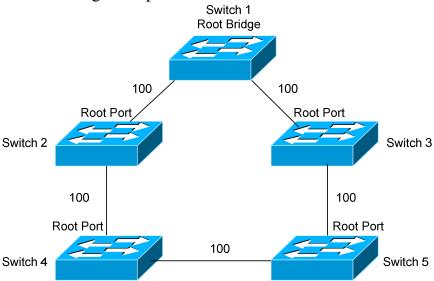
Identifying Root Ports

The second step in the STP process is identifying **Root Ports**, or the port on each switch that has the lowest **path cost** to get to the Root Bridge. Each switch has only **one** Root Port, and the Root Bridge *cannot* have a Root Port.

Path Cost is a cumulative cost based on the bandwidth of the links. The higher the bandwidth, the lower the Path Cost:

Bandwidth	Cost
4 Mbps	250
10 Mbps	100
16 Mbps	62
100 Mbps	19
1 Gbps	4

Consider the following example:



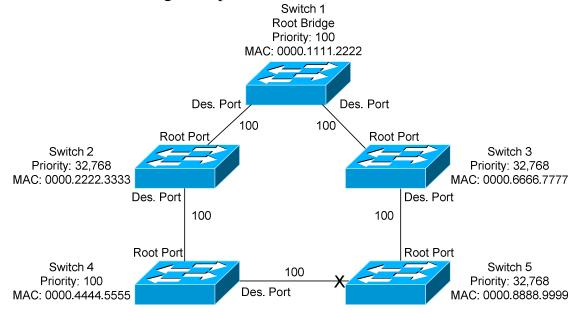
Assume the links between all switches are 10Mbps Ethernet, with a Path Cost of 100. Each switch will identify the port with the least cumulative Path Cost to get to the Root Bridge.

For Switch 4, the port leading up to Switch 2 has a Path Cost of 200, and becomes the Root Port. The port to Switch 5 has a higher Path Cost of 300. The Root Port is said to have received the most **superior BPDU** to the Root Bridge. Likewise, non-Root Ports are said to have received **inferior BPDU's** to the Root Bridge.

Identifying Designated Ports

The third and final step in the STP process is to identify **Designated Ports.** Each *network segment* requires a single Designated Port, which has the lowest path cost leading to the Root Bridge. This port *will not* be placed in a blocking state. A port cannot be both a Designated Port and a Root Port.

Consider the following example:



Ports on the Root Bridge are *never* placed in a blocking state, and thus become Designated Ports for directly attached segments.

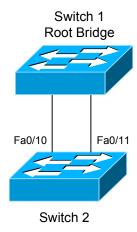
The network segments between Switches 2 and 4, and between Switches 3 and 5, both require a Designated Port. The ports on Switch 2 and Switch 3 have the lowest Path Cost to the Root Bridge for the two respective segments, and thus both become Designated Ports.

The segment between Switch 4 and Switch 5 does not contain a Root Port. One of the ports must be elected the Designated Port for that segment, and the other must be placed in a blocking state.

Normally, Path Cost is used to determine which port is blocked. However, the ports connecting Switches 4 and 5 have the same Path Cost to reach the Root Bridge (200). Whichever switch has the **lowest** Bridge ID is awarded the Designated Port. Whichever switch has the **highest** Bridge ID has its port placed in a blocking state. In this example, Switch 4 has the lowest priority, and thus Switch 5's port goes into a blocking state.

Port ID

In certain circumstances, a tie will occur in both Path Cost and Bridge ID. Consider the following example:



If the bandwidth of both links are equal, then both of Switch 2's interfaces have an equal path cost to the Root Bridge. Which interface will become the Root Port? The tiebreaker should be the lowest Bridge ID, but that cannot be used in this circumstance (unless Switch 2 has become schizophrenic).

In this circumstance, **Port ID** will be used as the tiebreaker. An interface's Port ID consists of two parts - a 6-bit **port priority** value, and the MAC address for that port. Whichever interface has the **lowest** Port ID will become the Root Port.

By default, the port priority of an interface is **128**. Lowering this value will ensure a specific interface becomes the Root Port:

Switch(config)# int fa0/10 **Switch(config-if)#** spanning-tree port-priority 50

Remember, that port priority is the *last* tiebreaker STP will consider. STP decides Root and Designated Ports based on the following criteria, and *in this order*:

- Lowest Path Cost to the Root Bridge
- Lowest Bridge ID
- Lowest Port ID

Extended System IDs

Normally, a switch's Bridge ID is a 64-bit value that consists of a 16-bit Bridge Priority value, and a 48-bit MAC address.

However, it is possible to include a VLAN ID, called an **extended System ID**, into a Bridge ID. Instead of *adding* bits to the existing Bridge ID, 12 bits of the Bridge Priority value are used for this System ID, which identifies the VLAN this STP process represents.

Because 12 bits have been stolen from the Bridge Priority field, the range of priorities has been reduced. Normally, the Bridge Priority can range from 0 (or *off*) to 65,535, with a default value of 32,768. With extended System ID enabled, the Priority range would be 0 - 61,440, and only in multiples of 4,096.

To enable the extended System ID:

Switch(config)# spanning-tree extend system-id

Enabling extended System ID accomplishes two things:

- Increases the amount of supported VLANs on the switch from 1005 to 4094.
- Includes the VLAN ID as part of the Bridge ID.

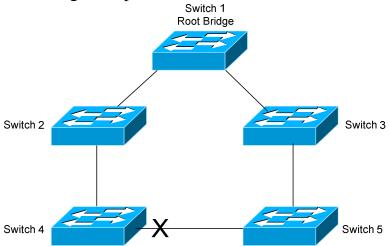
Thus, when this command is enabled, the 64-bit Bridge ID will consist of the following:

- 4-bit Priority Value
- 12-bit System ID value (VLAN ID)
- 48-bit MAC address

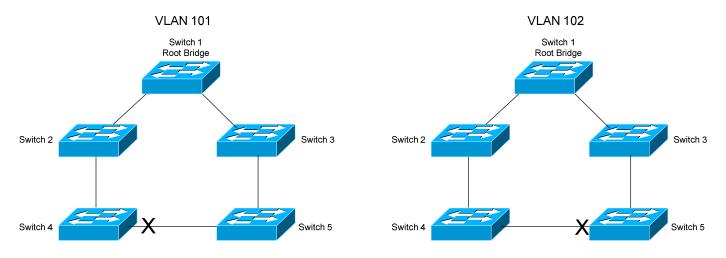
<u>Per-VLAN Spanning Tree (PVST) Example</u>

Remember that PVST+ is the default implementation of STP on Catalyst switches. Thus, each VLAN on the switch is allotted its own STP process.

Consider the following example:



With Common Spanning Tree (CST), all VLANS would belong to the same STP process. Thus, if one Switch 4's ports entered a *blocking* state to eliminate the loop, all VLANs would be blocked out that port. For efficiency purposes, this may not be ideal.



In the above examples, the benefit of PVST becomes apparent. STP runs a separate process for each VLAN, allowing a port to enter a blocking state **only for that specific VLAN.** Thus, it is possible to load balance VLANs, allowing traffic to flow more efficiently.

STP Port States

Switch ports participating in STP progress through five **port states**:

Blocking – The *default* state of an STP port when a switch is powered on, and when a port is shut down to eliminate a loop. Ports in a blocking state do not forward frames or learn MAC addresses. It will still listen for BPDUs from other switches, to learn about changes to the switching topology.

Listening – A port will progress from a Blocking to a Listening state only if the switch believes that the port *will not be shut down* to eliminate a loop. The port will listen for BPDU's to participate in the election of a Root Bridge, Root Ports, and Designated Ports. Ports in a listening state will not forward frames or learn MAC addresses.

Learning – After a brief period of time, called a **Forward Delay**, a port in a listening state will be elected either a Root Port or Designated Port, and placed in a learning state. Ports in a learning state listen for BPDUs, and also begin to learn MAC addresses. However, ports in a learning state will still not forward frames.

(Note: If a port in a listening state *is not* kept as a Root or a Designated Port, it will be placed into a *blocking* state and not a *learning* state.)

Forwarding – After another Forward Delay, a port in learning mode will be placed in forwarding mode. Ports in a forwarding state can send and receive all data frames, and continue to build the MAC address table. All designated, root, and non-uplink ports will eventually be placed in a forwarding state.

Disabled – A port in disabled state has been administratively shut down, and does not participate in STP or forward frames at all.

On average, a port in a blocking state will take 30 to 50 seconds to reach a forwarding state.

To view the current state of a port (such fa0/10):

Switch# show spanning-tree interface fa0/10

```
Interface Fa0/10 in Spanning tree 1 is Forwarding
Port path cost 100, Port priority 128
<snip>
```

(Reference: http://www.cisco.com/en/US/docs/switches/lan/catalyst4500/12.1/8aew/configuration/guide/spantree.html#wp1020487)

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STP Timers

STP utilizes three timers to ensure all switches remain synchronized, and to allow enough time for the Spanning Tree process to ensure a loop-free environment.

- **Hello Timer** Default is **2 seconds**. Indicates how often BPDU's are sent by switches.
- Forward Delay Default is 15 seconds. Indicates a *delay* period in both the *listening* and *learning* states of a port, for a total of 30 seconds. This delay ensures STP has ample time to detect and eliminate loops.
- Max Age Default is 20 seconds. Indicates how long a switch will keep BPDU information from a neighboring switch before discarding it. In other words, if a switch fails to receive BPDU's from a neighboring switch for the Max Age period, it will remove that switch's information from the STP topology database.

All timer values can be adjusted, and should *only* be adjusted on the Root Bridge. The Root Bridge will **propagate the changed timers** to all other switches participating in STP. Non-Root switches will ignore their locally configured timers.

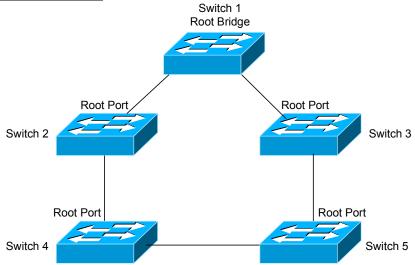
To adjust the three STP timers for VLAN 10:

Switch(config)# spanning-tree vlan 10 hello-time 10 Switch(config)# spanning-tree vlan 10 forward-time 20 Switch(config)# spanning-tree vlan 10 max-age 40

The timers are measured in seconds. The above examples represent the maximum value each timer can be configured to.

Remember that STP is configured on a VLAN by VLAN basis on Catalyst Switches.

STP Topology Changes



An **STP topology change** will occur under two circumstances:

- When an interface is placed into a Forwarding state.
- When an interface *already* in a Forwarding or Learning state is placed into a Blocking state.

The switch recognizing this topology change will send out a **TCN** (**Topology Change Notification**) BPDU, destined for the Root Bridge. The TCN BPDU does not contain any data about the actual change – it only indicates that a change occurred.

For example, if the interface on Switch 4 connecting to Switch 5 went down, Switch 4 would send a TCN out its Root Port to Switch 2. Switch 2 will acknowledge this TCN by sending a BPDU *back* to Switch 4 with the **Topology Change Acknowledgement (TCA) bit** set. Switch 2 would then forward the TCN out *its* Root Port to Switch 1 (the Root Bridge).

Once the Root Bridge receives the TCN, it will send out a BPDU with the **Topology Change (TC) bit** set to all switches. When a switch receives this Root BPDU, it will temporarily lower its MAC-address Aging Timer from 300 seconds to **15 seconds**, so that any erroneous MAC addresses can be quickly flushed out of the CAM table.

The MAC-Address Aging Timer will stay lowered to 15 seconds for a period of **35 seconds** by default, or one Max Age (20 seconds) plus one Forward Delay (15 seconds) timer.

(Reference: http://www.cisco.com/en/US/tech/tk389/tk621/technologies_tech_note09186a0080094797.shtml)

Basic STP Configuration

To disable STP for a specific VLAN:

Switch(config)# no spanning-tree vlan 10

To adjust the Bridge Priority of a switch from its default of 32,768, to increase its chances of being elected Root Bridge of a VLAN:

Switch(config)# spanning-tree vlan 10 priority 150

To change an interface's Path Cost from its defaults:

Switch(config)# int fa0/24 Switch(config-if)# spanning-tree cost 42

To force a switch to become the Root Bridge:

Switch(config)# spanning-tree vlan 10 root primary

The *root primary* parameter in the above command automatically lowers the switch's priority to 24,576. If another switch on the network has a lower priority than 24,576, the above command will lower the priority by 4096 less than the priority of the other switch.

It is possible to assign a **Secondary Root Bridge** for redundancy. To force a switch to become a Secondary Root Bridge:

Switch(config)# spanning-tree vlan 10 root secondary

The *root secondary* parameter in the above command automatically lowers the switch's priority to 28,672.

To specify the *diameter* of the switching topology:

Switch(config)# spanning-tree vlan 10 root primary diameter 7

The *diameter* parameter in the preceding command indicates the length of the STP topology (number of switches). The maximum (and default) value for the diameter is 7. Note that the switching topology can contain more than seven switches; however, each *branch* of the switching *tree* can only extend seven switches deep, from the Root Bridge.

The *diameter* command will also adjust the Hello, Forward Delay, and Max Age timers. This is the **recommended way** to adjust timers, as the hello timers are tuned specifically to the diameter of the switching network.

STP PortFast

PortFast allows switch ports that connect a host device (such as a printer or a workstation), to bypass the usual progression of STP states. Theoretically, a port connecting to a host device can never create a switching loop. Thus, Port Fast allows the interface to move from a *blocking* state to a *forwarding* state **immediately**, eliminating the normal 30 second STP delay.

To configure PortFast on an interface:

Switch(config)# int fa0/10 Switch(config-if)# spanning-tree portfast

To enable PortFast globally on all interfaces:

Switch(config)# spanning-tree portfast default

PortFast *should not* be enabled on switch ports connecting to another hub/switch, as this may result in a loop. Note that PortFast does *not* disable STP on an interface - it merely speeds up the convergence.

PortFast additionally **reduces unnecessary BPDU traffic**, as TCN BPDU's will not be sent out for state changes on a PortFast-enabled interface.

STP UplinkFast

Switches can have multiple uplinks to other *upstream* switches. If the multiple links are not placed in an EtherChannel, then at least one of the ports is placed into a *blocking* state to eliminate the loop.

If a **directly-connected** interface goes down, STP needs to perform a recalculation to bring the other interface out of a blocking state. As stated earlier, this calculation can take from 30 to 50 seconds.

UplinkFast allows the port in a *blocking* state to be held in standby-mode, and activated immediately if the *forwarding* interface fails. If multiple ports are in a *blocking* state, whichever port has the lowest Root Path Cost will become unblocked. The Root Bridge *cannot* have UplinkFast enabled.

UplinkFast is configured globally for all VLANs on the switch:

Switch(config)# spanning-tree uplinkfast

 $(Reference: \underline{http://www.cisco.com/en/US/docs/switches/lan/catalyst3750/software/release/12.2_35_se/configuration/guide/swstpopt.html}) \\$

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STP BackboneFast

While UplinkFast allows faster convergence if a *directly-connected* interface fails, **BackboneFast** provides the same benefit is an **indirectly-connected** interface fails.

For example, if the Root Bridge fails, another switch will be elected the Root. A switch learning about the new Root Bridge must wait its Max Age timer to flush out the old information, before it will accept the updated info. By default, the Max Age timer is 20 seconds.

BackboneFast allows a switch to bypass the Max Age timer if it detects an **indirect failure** on the network. It will update itself with the new Root info immediately.

BackboneFast is configured globally, and should be implemented on all switches in the network when used:

Switch(config)# spanning-tree backbonefast

Protecting STP

STP is vulnerable to attack for two reasons:

- STP builds its topology information by accepting a neighboring switch's BPDU's.
- The Root Bridge is always determined by the lowest Bridge ID.

Switches with a low priority can be maliciously placed on the network, and elected the Root Bridge. This may result in a suboptimal or unstable STP topology.

Cisco implemented three mechanisms to protect the STP topology:

- Root Guard
- BPDU Guard
- BPDU Filtering

All three mechanisms are configured on an individual interface basis, and are *disabled* by default. When enabled, these mechanisms apply to all VLANs for that particular interface.

 $(Reference: \underline{http://www.cisco.com/en/US/docs/switches/lan/catalyst3750/software/release/12.2\underline{\ 35\ se/configuration/guide/swstpopt.html})$

Root Guard

Root Guard prevents an unauthorized switch from advertising itself as a Root Bridge.

Switch(config)# interface fa0/10 **Switch(config-if)**# spanning-tree guard root

The above command will prevents the switch from accepting a new Root Bridge off of the fa0/10 interface. If a Root Bridge advertises itself to this port, the port will enter a *root-inconsistent* state (a pseudo-blocking state):

Switch# *show spanning-tree inconsistentports*

Name	Interface	Inconsistency
VLAN100	FastEthernet0/10	Root Inconsistent

BPDU Guard and BPDU Filtering

BPDU Guard is employed on interfaces that are PortFast-enabled. Under normal circumstances, a PortFast-enabled interface connects to a host device, and thus the interface should never receive a BPDU.

If another switch is accidentally or maliciously connected into a PortFast interface, BPDU Guard will place the interface into an **errdisable** state. More accurately, if an interface configured for BPDU Guard receives a BPDU, then the *errdisable* state will occur. To enable BPDU Guard:

Switch(config)# interface fa0/10 **Switch(config-if)#** spanning-tree bpduguard enable

To take an interface out of an errdisable state, simply disable and re-enable the interface:

Switch(config)# interface fa0/10 Switch(config-if)# shutdown Switch(config-if)# no shutdown

BPDU Filtering essentially disables STP on a particular interface, by preventing it from sending or receiving BPDU's:

Switch(config)# interface fa0/10 **Switch(config-if)**# spanning-tree bpdufilter enable

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<u>Unidirectional Link Detection (UDLD)</u>

Most communication in a switching network is bi-directional. STP requires that switches send BPDU's bi-directionally to build the topology database. If a malfunctioning switch port only allows traffic one way, and the switch still sees that port as up, a loop can form without the switch realizing it.

Unidirectional Link Detection (UDLD) periodically tests ports to ensure bi-directional communication is maintained. UDLD sends out **ID frames** on a port, and waits for the remote switch to respond with its own ID frame. If the remote switch does not respond, UDLD assumes the interface has malfunctioned and become unidirectional.

By default, UDLD sends out ID frames every **15 seconds**, and must be enabled on both sides of a link. UDLD can run in two modes:

- **Normal Mode** If a unidirectional link is detected, the port is *not* shut down, but merely flagged as being in an *undetermined* state
- **Aggressive Mode** If a unidirectional link is detected, the port is placed in an *errdisable* state

UDLD can be enabled globally (but only for Fiber ports on the switch):

```
Switch(config)# udld enable message time 20
Switch(config)# udld aggressive message time 20
```

The *enable* parameter sets UDLD into normal mode, and the *aggressive* parameter is for aggressive mode (obviously). The *message time* parameter modifies how often ID frames are sent out.

UDLD can be configured on individual interfaces:

```
Switch(config-if)# udld enable
Switch(config-if)# udld aggressive
Switch(config-if)# udld disable
```

To view UDLD status on ports, or re-enable UDLD errdisabled ports:

Switch# show udld Switch# udld reset

STP Troubleshooting Commands

To view STP information for a *specific* VLAN:

Switch# show spanning-tree vlan 100

```
VLAN0100
 Spanning tree enabled protocol ieee
 Root ID Priority 24576
           Address 00a.5678.90ab
           Cost
                    19
           Port 24 (FastEthernet0/24)
           Hello Time 2 sec Max Age 20 sec Forward Delay 15 sec
 Bridge ID Priority 32768 (priority 32768 sys-id-ext 1)
Address 000c.1234.abcd
           Hello Time 2 sec Max Age 20 sec Forward Delay 15 sec
           Aging Time 300
                                   Prio.Nbr
              Role Sts Cost
Interface
_____ ___
                    _____
Fa0/24
            Root FWD 19 128.24
Altn BLK 19 128.23
Fa0/23
```

To view STP information for all VLANS:

Switch# show spanning-tree

To view detailed STP interface information:

Switch# show spanning-tree detail

VLAN100 is executing the ieee compatible Spanning Tree protocol Bridge Identifier has priority 32768, address 000c.1234.abcd Configured hello time 2, max age 20, forward delay 15

<snip>

Port 23 (FastEthernet0/23) of VLAN100 is forwarding Port path cost 19, Port priority 128, Port Identifier 128.23. Designated root has priority 24576, address 00a.5678.90ab Designated bridge has priority 24576, address 00a.5678.90ab Designated port id is 128.23, designated path cost 0

<snip>

 $(Reference: \underline{http://www.cisco.com/en/US/docs/switches/lan/catalyst6500/ios/12.1E/native/command/reference/show4.\underline{html\#wp1026768})) + \underline{http://www.cisco.com/en/US/docs/switches/lan/catalyst6500/ios/12.1E/native/command/reference/show4.\underline{html\#wp1026768})) + \underline{http://www.cisco.com/en/US/docs/switches/lan/catalyst6500/ios/12.1E/native/command/reference/show4.\underline{html#wp1026768})) + \underline{http://www.cisco.com/en/US/docs/switches/lan/catalyst6500/ios/12.1E/native/command/reference/show4.\underline{html#wp1026768})) + \underline{http://www.cisco.com/en/US/docs/switches/lan/catalyst6500/ios/12.1E/native/command/reference/show4.\underline{html#wp1026768})) + \underline{http://www.cisco.com/en/US/docs/switches/lan/catalyst6500/ios/12.1E/native/command/reference/show4.\underline{html#wp1026768})) + \underline{http://www.cisco.com/en/US/docs/switches/lan/catalyst6500/ios/12.1E/native/command/reference/show4.\underline{html#wp1026768}) + \underline{http://www.cisco.com/en/US/docs/switches/lan/catalyst6500/ios/12.\underline{html#wp1026768}) + \underline{html#wp1026768} + \underline{html#wp1026768}) + \underline{html#wp1026768} + \underline{html#wp1026768} + \underline{html#wp1026768} + \underline{html#$

Rapid Spanning Tree Protocol (RSTP)

To further alleviate the 30 to 50 second convergence delays with STP, enhancements were made to the original IEEE 802.1D standard. The result was **802.1w**, or **Rapid Spanning Tree Protocol (RSTP)**.

RSTP is similar in many respects to STP. BPDU's are forwarded between switches, and a Root Bridge is elected, based on the lowest Bridge ID. Root Ports and Designated Ports are also elected. RSTP defines five **port types**:

- **Root Port** Switch port on each switch that has the best Path Cost to the Root Bridge (same as STP).
- Alternate Port A backup Root Port, that has a less desirable Path Cost. An Alternate Port is placed in a *discarding* state.
- **Designated Port** Non-Root port that represents the best Path Cost for each network segment to the Root Bridge (same as STP). Designated ports are also referred to as **Point-to-Point** ports.
- **Backup Port** A backup Designated Port, that has a less desirable Path Cost. A Backup Port is placed in a *discarding* state.
- Edge Port A port connecting a host device, which is moved to a Forwarding state immediately. If an Edge Port receives a BPDU, it will lose its Edge Port status and participate in RSTP calculations. On Cisco Catalyst switches, any port configured with PortFast becomes an Edge Port.

The key benefit of RSTP is speedier convergence. Switches no longer require artificial Forwarding Delay timers to ensure a loop-free environment.

Switches instead perform a **handshake synchronization** to ensure a consistent topology table. During initial convergence, the Root Bridge and its directly-connected switches will place their interfaces in a *discarding* state. The Root Bridge and those switches will exchange BPDU's, synchronize their topology tables, and then place their interfaces in a forwarding state.

Each switch will then perform the same handshaking process with their downstream neighbors. The result is convergence that completes in a few seconds, as opposed to 30 to 50 seconds.

(Reference: http://www.cisco.com/en/US/tech/tk389/tk621/technologies_white_paper09186a0080094cfa.shtml)

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Rapid Spanning Tree Protocol (RSTP) (continued)

Changes to the RSTP topology are also handled more efficiently than 802.1D STP.

Recall in that in 802.1D STP, a switch recognizing a topology change will send out a **TCN** (**Topology Change Notification**) BPDU, destined for the Root Bridge. Once the Root Bridge receives the TCN, it will send out a BPDU with the **Topology Change** (**TC**) bit set to all switches. When a switch receives this Root BPDU, it will temporarily lower its MAC-address Aging Timer from 300 seconds to **15 seconds**, so that any erroneous MAC addresses can be quickly flushed out of the CAM table.

In RSTP, a switch recognizing a topology change *does not* have to inform the Root Bridge first. Any switch can generate and forward a **TC BPDU**. A switch receiving a TC BPDU will **flush all MAC addresses learned on all ports**, except for the port that received the TC BPDU.

RSTP incorporates the features of UplinkFast by allowing Alternate and Backup ports to immediately enter a Forwarding state, if the primary Root or Designated port fails. RSTP also inherently employs the principles of BackboneFast, by not requiring an arbitrary Max Age timer for accepting inferior BPDU's if there is an indirect network failure.

802.1w RSTP is backwards-compatible with 802.1D STP. However, when RSTP switches interact with STP switches, RSTP loses its inherent advantages, as will perform according to 802.1D specifications.

Two separate standards of RSTP have been developed:

- Rapid Per-VLAN Spanning Tree Protocol (RPVST+) Cisco's proprietary implementation of RSTP.
- Multiple Spanning Tree (MST) The IEEE 802.1s standard or RSTP.

(Reference: http://www.cisco.com/en/US/tech/tk389/tk621/technologies_white_paper09186a0080094cfa.shtml)

Multiple Spanning Tree (MST)

Earlier in this guide, two types of STP were defined:

- Common Spanning Tree (CST) All VLANs utilize one STP process
- Per-VLAN Spanning Tree (PVST) Each VLAN is allotted its own STP process

PVST allows for more efficient traffic flow throughout the switching network. However, each VLAN must run its own *separate* STP process, often placing an extreme burden on the switch's processor.

Multiple Spanning Tree (MST) allows *groups* of VLANs to be allotted their own STP process. Each STP process is called an **instance.** MST separates the STP topology into **regions** that must contain identical parameters, including:

- Configuration Name a 32-bit value similar to a VTP domain
- **Revision Number** a 16-bit value that identifies the current MST configuration's revision.
- VLAN-to-Instance Mappings

Each **region** runs its own **Internal Spanning Tree** (**IST**) to eliminate loops within that region. IST is essentially an enhanced form of RSTP that supports MST-specific parameters.

MST is fully compatible with all other implementations of STP.

(Reference: http://www.cisco.com/en/US/docs/switches/lan/catalyst4500/12.2/31sg/configuration/guide/spantree.pdf)

MST Configuration

MST must first be enabled globally on a switch:

Switch(config)# spanning-tree mode mst

Most other MST configuration is completed in "MST Configuration" mode:

Switch(config)# spanning-tree mst configuration

To configure the switch's MST Configuration Name:

Switch(config-mst)# name MYMSTNAME

To configure the switch's Revision Number:

Switch(config-mst)# revision 10

To map VLANs to a specific MST instance:

Switch(config-mst)# instance 2 vlan 1-100

A maximum of **16 instances** are allowed (0-15). By default, all VLANs belong to **instance 0**. Recall that the above three parameters (configuration name, revision number, and mappings) must be identical on all MST switches in a region.

To view the changes to the configuration:

Switch(config-mst)# show pending

```
Pending MST configuration
Name [MYMSTNAME]
Revision 10
Instance Vlans mapped
-----
0 101-4094
2 1-100
```

All other configuration of MST is identical to standard STP, with two exceptions. The parameter "*mst*" must be used, and all settings are applied to **instances** instead of **VLANs**.

Switch(config)# spanning-tree mst 2 root primary **Switch(config)**# spanning-tree mst 2 priority 32000

The above two configurations are applied to MST Instance 2.

Section 6 - IPv4 Addressing and Subnetting -

Hardware Addressing

The **hardware address** is used by devices to communicate on the *local* network. Hardware addressing is a function of the **data-link** layer of the OSI model (Layer-2).

The hardware address for Ethernet networks is the MAC address, a 48-bit hexadecimal address that is usually hard-coded on the network card. In theory, this means the MAC address cannot be altered; however, the MAC address is often stored in flash on the NIC, and thus can be changed with special utilities.

MAC addresses can be represented in two formats (either notation is acceptable):

00:43:AB:F2:32:13 0043.ABF2.3213

The MAC address has one shortcoming – *it contains no hierarchy*. There is **no mechanism** to create **boundaries** between networks.

Instead, the first six hexadecimal digits of a MAC identify the manufacturer of the network card (referred to as the **OUI** (**Organizational Unique Identifier**)), and the last 6 digits identify the host device (referred to as the **host ID**). Still, there is no way to distinguish one *network* from another.

Imagine the difficulties this poses. If *only* hardware addressing existed, all devices would technically be on the *same* network. Modern internetwork systems like the Internet could not exist, as there would be no way to separate *my* network from *your* network.

Furthermore, imagine if the entire Internet was a purely switched, data-link layer environment. Switches, as a rule, forward broadcasts out *all* ports. Guesstimating that there are billions devices on the Internet, with each device sending out a broadcast on average every few seconds, the resulting broadcast storms would be devastating. The Internet would simply collapse.

The need for **logical addressing**, and **routers**, became apparent.

Logical Addressing

Logical addressing is a function of the **network** layer of the OSI Model (Layer-3).

Logical addresses, unlike hardware addresses, provide a hierarchical structure to separate networks. A logical address identifies not only a unique **Host ID**, but also the **network** that host belongs to. Additionally, logical addresses are rarely hard-coded onto hosts, and can be changed freely.

Two common logical addressing protocols are **IPX** (**Internetwork Packet Exchange**) and **IP** (**Internet Protocol**). IPX was predominantly used on Novell networks, but is mostly deprecated. **IP** is the most widely-used logical address today.

Internet Protocol (IP)

IP was developed by the Department of Defense (DoD) during the late 1970's. It was included in a group of protocols that became known as the **TCP/IP protocol suite**.

The DoD developed their own networking model to organize and define the TCP/IP protocol suite. This became known as the **DoD Model**, and consists of four layers:

	OSI Model		DoD Model	Example Protocols
7	Application			
6	Presentation	4	Application	FTP, HTTP, SMTP
5	Session			
4	Transport	3	Host-to-Host	TCP, UDP
3	Network	2	Internet	IP
2	Data-link	1	Network Access	Ethernet
1	Physical	1	Network Access	Ethernet

IP provides two core functions:

- Logical addressing of hosts
- **Routing** of packets between networks.

IP has undergone several revisions. IP Version 4 (**IPv4**) is currently in widespread deployment, but will eventually be replaced with IP Version 6 (**IPv6**). This guide will concentrate on IPv4, and IPv6 will be covered extensively in a separate guide.

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IPv4 Addressing

One of IP's core functions is to provide logical addressing for hosts. An **IP** address provides a hierarchical structure to separate networks. Consider the following address as an example:

158.80.164.3

An IP address is separated into four **octets**:

First Octet	Second Octet	Third Octet	Fourth Octet
158	.80	.164	.3

Each octet is 8 bits long, resulting in a **32-bit IP address**. A computer understands an IP address in its binary form; the above address in binary would look as follows:

First Octet	Second Octet	Third Octet	Fourth Octet
10011110	.01010000	.10100100	.00000011

Part of the above IP address identifies the *network*. The other part of the address identifies the *host*. A **subnet mask** helps make this distinction. Consider the following:

158.80.164.3 255.255.0.0

The above IP address has a subnet mask of 255.255.0.0. The subnet mask follows two rules:

- If a binary bit is set to a **1** (or **on**) in a subnet mask, the corresponding bit in the address identifies the **network**.
- If a binary bit is set to a **0** (or **off**) in a subnet mask, the corresponding bit in the address identifies the **host**.

Looking at the above address and subnet mask in binary:

Address:	10011110.01010000.10100100.00000011
Subnet Mask:	11111111.111111111.00000000.00000000

The first 16 bits of the subnet mask are set to 1. Thus, the first 16 bits of the address (158.80) identify the *network*. The last 16 bits of the subnet mask are set to 0. Thus, the last 16 bits of the address (164.3) identify the unique *host* on that network.

IPv4 Addressing (continued)

Hosts on the same logical network will have *identical* network addresses, and can communicate freely. For example, the following two hosts are on the same network:

Host A: 158.80.164.100 255.255.0.0 Host B: 158.80.164.101 255.255.0.0

Both share the same network address (158.80), which is determined by the 255.255.0.0 subnet mask. Hosts that are on *different* networks cannot communicate without an intermediating device. For example:

Host A: 158.80.164.100 255.255.0.0 Host B: 158.85.164.101 255.255.0.0

The subnet mask has remained the same, but the network addresses are now different (158.80 and 158.85 respectively). Thus, the two hosts are *not* on the same network, and cannot communicate without a **router** between them. **Routing** is the process of sending packets from one network to another.

Consider the following, trickier example:

Host A: 158.80.1.1 255.248.0.0 Host B: 158.79.1.1 255.248.0.0

The specified subnet mask is now 255.248.0.0, which doesn't fall cleanly on an octet boundary. To determine if these hosts are on separate networks, first convert everything to binary:

Host A Address: 10011110.01010000.00000001.00000001
Host B Address: 10011110.01001111.00000001.00000001
Subnet Mask: 11111111.11111000.00000000.00000000

Remember, the **1** (or **on**) bits in the subnet mask identify the *network* portion of the address. In this example, the first *13 bits* (the 8 bits of the first octet, and the first 5 bits of the second octet) identify the network. Looking at only the first 13 bits of each address:

Host A Address: 10011110.01010 Host B Address: 10011110.01001

Clearly, the network addresses are *not* identical. Thus, these two devices are on separate networks, and require a router to communicate.

IP Address Classes

The IPv4 address space has been structured into several **classes**. The value of the **first octet** of an address determines the class of the network:

Class	First Octet Range	Default Subnet Mask
Class A	1 - 127	255.0.0.0
Class B	128 - 191	255.255.0.0
Class C	192 - 223	255.255.255.0

Class A networks range from 1 to 127. The *default* subnet mask is 255.0.0.0; thus, by *default*, the first octet defines the network, and last three octets define the host. This results in a maximum of 127 Class A networks, with 16,777,214 hosts per network!

Example of a Class A address:

Address: 64.32.254.100 Subnet Mask: 255.0.0.0

Class B networks range from 128 to 191. The *default* subnet mask is 255.255.0.0; thus, by *default*, the first two octets define the network, and the last two octets define the host. This results in a maximum of 16,384 Class B networks, with 65,534 hosts per network.

Example of a Class B address:

Address: 152.4.12.195 Subnet Mask: 255.255.0.0

Class C networks range from **192** to **223.** The default subnet mask is 255.255.255.0; thus, by *default*, the first three octets define the network, and the last octet defines the host. This results in a maximum of **2,097,152** Class C networks, with **254** hosts per network.

Example of a Class C address:

Address: 207.79.233.6 Subnet Mask: 255.255.255.0

CIDR (Classless Inter-Domain Routing)

Classless Inter-Domain Routing (CIDR) is simplified method of representing a subnet mask. CIDR identifies the number of binary bits set to a 1 (or on) in a subnet mask, preceded by a slash.

Consider the following subnet mask: 255.255.255.240

Looking at the above subnet mask in binary:

11111111.111111111.1111111111.11110000

The first 28 bits of the above subnet mask are set to 1. To represent this in CIDR notation: /28

Consider this next example:

192.168.1.1 255.255.255.0

The above address/subnet mask can be represented as follows using CIDR:

192.168.1.1 /24

Address "Classes" vs. Subnet Mask

Remember the following three rules:

- The **first octet** on an address dictates the *class* of that address.
- The **subnet mask** dictates what portion of an address identifies the network, and what portion identifies the host.
- Each class has a *default* subnet mask.

Thus, the address 10.1.1.1 is a Class A address, and its default subnet mask is 255.0.0.0 (or in CIDR, /8). However, it is possible to use subnet masks *other* than the default, such as applying a Class B mask to a Class A address:

10.1.1.1 /16

However, **this does not change the class of the above address**. It remains a Class A *address*, which has been subnetted using a Class B *mask*.

Remember, the *only* thing that determines the class of an IP address is the first octet of that address. Likewise, the subnet mask is the *only* thing that determines what portion of an address is the network, and which portion is the host.

Subnet and Broadcast Addresses

Two addresses have been reserved on each network for special use. Each network must have a **subnet** (or **network**) address, and a **broadcast** address.

Neither of these addresses can be assigned to a host device.

The **subnet** address is used to *identify the network* itself. Routing tables contain lists of networks, and each network is identified by its subnet address. Subnet addresses contain **all 0 bits in the host portion** of the address.

For example, the following is a subnet address: 192.168.1.0/24

The **broadcast** address identifies *all* hosts on a particular network. A packet sent to the broadcast address will be received and processed by every device on that network. Broadcast addresses contain **all 1 bits in the host portion** of the address.

For example, the following is a broadcast address: 192.168.1.255/24

Broadcasts are one of three types of IP packets:

- Unicasts are packets sent from one host to another host
- **Multicasts** are packets sent from one host to a *group* of hosts
- **Broadcasts**, as stated earlier, are packets sent from one host to all other hosts on the local network

A router, by default, will **never forward** a multicast or broadcast packet from one interface to another.

A switch, be default, will forward a multicast or broadcast **out every port**, except for the port that sent the multicast/broadcast.

Subnetting

Subnetting is the process of creating new networks (or *subnets*) by **stealing bits** from the host portion of a subnet mask. There is one caveat: stealing bits from hosts creates **more** networks but **fewer** hosts per network. Thus, every time a network is subnetted, addresses are lost.

Consider the following Class C network:

The default subnet mask for this network is 255.255.25.0. This single network can be segmented, or *subnetted*, into multiple networks. For example, assume a minimum of *10* new networks are required. Resolving this is possible using the following magical formula:

$$2^{n}-2$$

The exponent 'n' identifies the number of bits to steal from the host portion of the subnet mask. The default Class C mask (255.255.255.0) looks as follows in binary:

11111111.11111111.1111111.00000000

There are a total of 24 bits set to 1, which are used to identify the network. There are a total of 8 bits set to 0, which are used to identify the host, and these host bits can be 'stolen.'

Stealing bits essentially involves changing host bits (set to 0 or off) in the subnet mask to network bits (set to 1 or on). Network bits in a subnet mask **must always be sequential**, skipping bits is not allowed.

Consider the result if three bits are stolen. Using the above formula:

$$2^{n}-2 = 2^{3}-2 = 8-2 = 6$$
 new networks created

However, a total of six new networks *does not* meet the original requirement of at least 10 networks. Consider the result if four bits are stolen:

$$2^{n}-2 = 2^{4}-2 = 16-2 = 14$$
 new networks created

A total of fourteen new networks *does* meet the original requirement. Stealing four host bits results in the following *new* subnet mask:

<u>Subnetting (continued)</u>

In the previous example, a Class C network was subnetted to create 14 new networks, using a subnet mask of 255.255.255.240 (or /28 in CIDR). Four bits were stolen in the subnet mask, leaving only four bits for hosts.

To determine the number of hosts this results in, for each of the new 14 networks, the same formula can be used: $2^n - 2$

Consider the result if four bits are available for hosts:

$$2^{n}-2 = 2^{4}-2 = 16-2 = 14$$
 usable hosts per network

Thus, subnetting a Class C network with a /28 mask creates fourteen new networks, with fourteen usable hosts per network.

The "-2" Rule of Subnetting

There is a specific purpose for the '-2' portion of the $2^n - 2$ formula. Previously, it was unacceptable to use an address that contained all '0' or all '1' bits in the *network* portion of the address.

However, this is **no longer true** on modern systems. Specifically, on Cisco IOS devices, the following command is now enabled by default:

The *ip subnet-zero* commands allows for the use of networks with all **'0'** or all **'1'** bits in the network portion of the address. Thus, the formula for calculating the number of new network is slightly altered, to simply **2**ⁿ. Consider if four bits are stolen for networks:

$$2^n = 2^4 = 16$$
 new networks created

However, it is **never possible to assign an address with all '0' or all '1' bits in the** *host* **portion of the address.** These are reserved for the subnet and broadcast addresses, respectively. Thus, the formula for calculating usable hosts is *always* $2^n - 2$.

Some have questioned whether CCNA/CCNP simulations and questions have *ip subnet-zero* enabled. It is generally accepted that having this enabled is now default behavior, and test questions should be answered accordingly. All future examples in this guide will assume the command is enabled.

Determining the 'Range' of Subnetted Networks

Determining the *range* of the newly created networks can be accomplished using several methods. The 'long' method involves some binary magic.

Still looking at the example 192.168.254.0 network, which was subnetted using a 255.255.255.240 mask:

192.168.254.0:	11000000.10101000.111111110.00000000
255.255.255.240:	11111111.111111111.11111111111110000

Subnetting stole four bits in the fourth octet, creating a total of 16 new networks (assuming *ip subnet-zero* is enabled). Looking at *only* the fourth octet, the first newly created network is 0000. The second new network is 0001. Calculating all possible permutations of the four stolen bits:

Binary	Decimal	Binary	Decimal	Binary	Decimal
.0000 <i>xxxx</i>	.0	.0110 <i>xxxx</i>	.96	.1100 <i>xxxx</i>	.192
.0001 <i>xxxx</i>	.16	.0111 <i>xxxx</i>	.112	.1101 <i>xxxx</i>	.208
.0010 <i>xxxx</i>	.32	.1000 <i>xxxx</i>	.128	.1110 <i>xxxx</i>	.224
.0011 <i>xxxx</i>	.48	.1001 <i>xxxx</i>	.144	.1111 <i>xxxx</i>	.240
.0100 <i>xxxx</i>	.64	.1010 <i>xxxx</i>	.160		
.0101 <i>xxxx</i>	.80	.1011 <i>xxxx</i>	.176		

Note that this equates to exactly 16 new networks. The decimal value represents the first (or the *subnet*) address of each newly created network. To determine the range for the hosts of the *first* new network:

Binary	Decimal	Binary	Decimal	Binary	Decimal
.0000 0000	.0	.0000 0110	.6	.0000 1100	.12
.0000 0001	.1	.0000 0111	.7	.0000 1101	.13
.0000 0010	.2	.0000 1000	.8	.0000 1110	.14
.0000 0011	.3	.0000 1001	.9	.0000 1111	.15
.0000 0100	.4	.0000 1010	.10		
.0000 0101	.5	.0000 1011	.11		

The binary value has been 'split' to emphasize the separation of the *network* bits from the *host* bits. The first address has all 0 bits in the host portion (0000), and thus is the **subnet address** for this network. The last address has all 1 bits in the host portion, and thus is the **broadcast address** for this network. Note that there are exactly **14 usable addresses** to assign to hosts.

Determining the 'Range' of Subnetted Networks (continued)

Calculating the ranges of subnetted networks can quickly become tedious when using the 'long' binary method. The 'shortcut' method involves taking the subnet mask (255.255.255.240 from the previous example), and subtracting the subnetted octet (240) from 256.

$$256 - 240 = 16$$

Assuming *ip subnet-zero* is enabled, the first network will begin at 0. Then, simply continue adding 16 to list the first address of each new network:

 $0 \quad 16 \quad 32 \quad 48 \quad 64 \quad 80 \quad 96 \quad 112 \quad 128 \quad 144 \quad 160 \quad 176 \quad 192 \quad 208 \quad 224 \quad 240$

Knowing the *first* address of each new network makes it simple to determine the *last* address of each network:

First address of network	0	16	32	48	64	80	96	112	128	144	
Last address of network	15	31	47	63	79	95	111	127	143	159	

Only the first 10 networks were calculated, for brevity. The first address of each network becomes the **subnet address** for that network. The last address of each network becomes the **broadcast address** for that network.

Once the first and last address of each network is known, determining the usable range for hosts is straightforward:

Subnet address	0	16	32	48	64	80	96	112	128	144	
	1	17	33	49	65	81	97	113	129	145	
Usable Range	\uparrow		1							1	
	14	30	46	62	78	94	110	126	142	158	
Broadcast address	15	31	47	63	79	95	111	127	143	159	

Hosts on the same network (such as 192.168.254.2 and 192.168.254.14) can communicate freely.

Hosts on different networks (such as 192.168.254.61 and 192.168.254.66) require a router to communicate.

Class A Subnetting Example

Consider the following subnetted Class A network: 10.0.0.0 255.255.248.0

Now consider the following questions:

- How many new networks were created?
- How many usable hosts are there per network?
- What is the full range of the first three networks?

By default, the 10.0.0.0 network has a subnet mask of 255.0.0.0. To determine the number of bits stolen:

255.0.0.0:	11111111.00000000.00000000.00000000
255.255.248.0:	11111111.111111111.11111000.00000000

Clearly, 13 bits have been stolen to create the new subnet mask. To calculate the total number of new networks:

$$2^n = 2^{13} = 8192$$
 new networks created

There are clearly **11 bits** remaining in the host portion of the mask:

$$2^{n}-2 = 2^{11}-2 = 2048-2 = 2046$$
 usable hosts per network

Calculating the ranges is a bit tricky. Using the 'shortcut' method, subtract the third octet (248) of the subnet mask (255.255.248.0) from 256.

$$256 - 248 = 8$$

The first network will begin at θ , again. **However**, the ranges are spread across multiple octets. The ranges of the first three networks look as follows:

Subnet address	10.0.0.0	10.0.8.0	10.0.16.0
	10.0.0.1	10.0.8.1	10.0.16.1
Usable Range	1		
	10.0.7.254	10.0.15.254	10.0.23.254
Broadcast address	10.0.7.255	10.0.15.255	10.0.23.255

Private vs Public Addresses

The rapid growth of the Internet resulted in a shortage of IPv4 addresses. In response, the powers that be designated a specific subset of the IPv4 address space to be *private*, to temporarily alleviate this problem.

A **public address** can be routed on the Internet. Thus, devices that should be Internet accessible (such as web or email servers) must be configured with public addresses.

A **private address** is only intended for use within an organization, and can never be routed on the internet. Three private addressing ranges were allocated, one for each IPv4 class:

- Class A 10.x.x.x
- Class B 172.16-31.x.x
- Class C 192.168.x.x

NAT (**Network Address Translation**) is used to translate between private addresses and public addresses. NAT allows devices configured with a private address to be *stamped* with a public address, thus allowing those devices to communicate across the Internet. NAT is covered in-depth in another guide.

NAT is only a temporarily solution to the address shortage problem. Eventually, IPv4 will be replaced with IPv6. This also is covered extensively in another guide.

Two other ranges, while not considered "private," have been reserved for specific use:

- 127.x.x.x reserved for diagnostic purposes. One such address (127.0.0.1), identifies the local host, and is referred to as the loopback or localhost address.
- **169.254.x.x** reserved for Automatic Private IP Addressing (APIPA). A host assigns itself an APIPA address if a DHCP server is unavailable to dynamically assign an address.

The IPv4 Header

The IPv4 header has **12 required fields** and **1 optional field**, and is 160 **bits** long.

Field	Length	Description
Version	4 bits	Version of IP (in this case, IPv4)
Header Length	4 bits	Specifies the length of the IP header (minimum 160 bits)
Type of Service	8 bits	Classifies traffic for QoS
Total Length	16 bits	Specifies the length of both the header and data payload
Identification	16 bits	Uniquely identifies fragments of a packet
Flags	3 bits	Flags for fragmentation
Fragment Offset	13 bits	Identifies the location of a fragment in a packet
Time to Live	8 bits	Decremented by each router traversed
Protocol	8 bits	Specifies the next upper layer protocol
Header Checksum	16 bits	Checksum for error checking
Source Address	32 bits	Source IPv4 address
Destination Address	32 bits	Destination IPv4 address
Options	32 bits	Optional field for various parameters

The **Identification, Flags,** and **Fragment Offset** fields are used in conjunction with each other. An IP packet larger than the MTU size of a link must be fragmented. Each fragment of the packet is marked with the same Identification number. The Fragment Offset allows the destination device to reassemble the fragments in the proper order.

The Flags field can dictate two conditions:

- **Don't Fragment (DF)** indicates the packet cannot be fragmented. If the packet reaches a link with a small MTU, the packet is then dropped, and an ICMP error message is sent back to the source.
- More Fragments (MF) all fragments have this bit set to one, *except* for the last fragment, where the bit is set to zero. This allows the destination device to know it has received all fragments.

IPv4 Protocol Numbers

The *Next Header* field is of some importance. This field identifies the next upper-layer header (for example, UDP, TCP or ICMP). These upper layer protocols are identified using IP Protocol Numbers.

The following is a list of common IP Protocol Numbers:

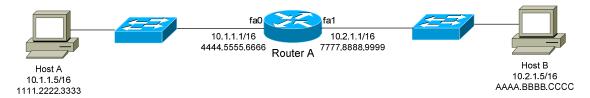
Protocol Number	Upper-Layer Protocol	
1	ICMP	
2	IGMP	
6	TCP	
9	IGRP	
17	UDP	
46	RSVP	
47	GRE	
50	IPSEC ESP	
51	IPSEC AH	
88	EIGRP	
89	OSPF	

(Reference: http://www.iana.org/assignments/protocol-numbers)

Resolving Logical Addresses to Hardware Addresses

Hosts cannot directly send data to another device's logical address. Network communication occurs across the data-link layer, using hardware addresses.

A mechanism is required to *map* logical addresses to hardware addresses. When using IP over an Ethernet network, the **Address Resolution Protocol** (**ARP**) provides this function for us. ARP allows a host to determine the MAC (hardware) address for a particular IP (logical) address.



Observe the above diagram. Following the step-by-step path a packet travels from HostA to the 10.2.1.5 address (HostB):

- First, HostA determines if the 10.2.1.5 address is *itself*. If the address is configured on a local interface, the packet never leaves HostA.
- Second, HostA determines if the 10.2.1.5 address is on the *same network* (or subnet). If it is, HostA will broadcast an ARP request, and wait for the appropriate host to reply with its MAC address.
- HostA determines that the 10.2.1.5 address is indeed on a *separate network*. It now parses its local routing table for a route to this remote network. Usually, hosts will be equipped with a **default route** (or, **default** *gateway*), to reach all other networks.
- Host A determines that RouterA is its default gateway. The host broadcasts an ARP request for RouterA's MAC address, and then forwards the packet to RouterA's MAC (4444.5555.6666).
- RouterA receives the packet, and parses at its own routing table. It determines that the 10.2.x.x network is directly attached off of its *fa1* interface. The router then broadcasts an ARP request for the 10.2.1.5 address.
- HostB responds to the router's ARP request with its MAC address (AAAA.BBBB.CCCC). RouterA is then able to forward the packet to HostB.

Troubleshooting IP using ICMP

Internet Control Message Protocol (ICMP) is used for a multitude of informational and error messaging purposes.

The following is a list of common ICMP types and codes:

Type	Code	Description
0	0	Echo Reply
3	_	Destination Unreachable
	0	Network Unreachable
	1	Host Unreachable
	2	Protocol Unreachable
	3	Port Unreachable
	4	Fragmentation Needed – Don't Fragment Flag Set
	6	Destination Network Unknown
	7	Destination Host Unknown
	9	Destination Network Administratively Prohibited
	10	Destination Host Administratively Prohibited
5		Redirect
8		Echo
11		TTL Exceeded

Several IP troubleshooting tools utilize ICMP, including **Packet Internet Groper (ping)** and **traceroute**.

Ping utilizes the Echo Request and Echo Reply ICMP messages to determine if a host is responding on a particular address.

Traceroute determines the routing path a packet takes to reach its destination.

Section 7 - TCP and UDP -

Transport Layer Protocols

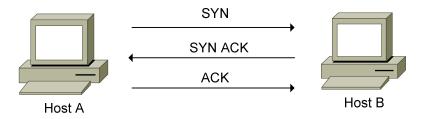
The **Transport** layer of the OSI model (or, the **Host-to-Host** layer of the DoD model) is concerned with the reliable transfer of data between devices. It ensures (or in some cases, does *not* ensure) that a packet arrives at its destination without corruption or data loss.

However, protocols at the transport layer do not actually *send* or *route* packets. Network layer protocols, such as IP, route packets from one network to another. In the TCP/IP protocol suite, TCP and UDP are transport layer protocols.

<u>Transmission Control Protocol (TCP)</u>

The **Transmission Control Protocol** (**TCP**) is defined as a reliable, **connection-oriented** transport protocol. Parameters must be agreed upon by both parties before a connection is established.

TCP utilizes a **three-way handshake** to accomplish this. Control messages are passed between two devices as the connection is set up:

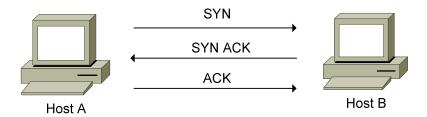


- Host A sends a **SYN** (short for **synchronize**) message to Host B to initiate a connection
- Host B responds with an ACK (short for acknowledgement) to Host A's SYN message, and sends its own SYN message (both messages are combined to form a SYN+ACK)
- Host A completes the handshake by sending an **ACK** to Host B's SYN.

The TCP header contains both a SYN flag and an ACK flag. Thus, when a particular message needs to be sent, the appropriate flag is marked as on (in other words, changed from a "0" to a "1"). A SYN+ACK message has both flags set to on (1).

<u>Transmission Control Protocol (TCP) (continued)</u>

Additionally, TCP **segments** data into smaller pieces for transport. Segments are assigned a sequence number, so that the receiving device can then reassemble this data in order upon arrival.



- Host A sends an **initial sequence number (ISN)** with its SYN message. This number is chosen from a random timer we'll assume an ISN of 4000.
- Host B responds to this sequence number with an acknowledgment number, which is always one more than the sequence number. Thus, Host B's acknowledgment number is 4001.
- Additionally, Host B sends an initial sequence number with *its* SYN message. We'll assume Host B's ISN is 6000.
- Host A responds to this sequence with an acknowledgement number of 6001.

After a TCP connection is established, each segment is tagged with a sequence number. TCP detects that a segment has been lost when it does *not* receive a corresponding acknowledgement of receipt. It must not only receive an **ACK**, but it must receive an ACK with the appropriate acknowledgement number.

(Reference: http://www.tcpipguide.com/free/t_TCPConnectionEstablishmentSequenceNumberSynchroniz.htm)

Additionally, TCP incorporates **windowing** for **flow control**. When flow control is enabled, both the sending and receiving devices must agree on the amount of data being sent in between acknowledgements. This helps prevent data loss due to one side of the connection being overloaded.

(Reference: http://www.tcpipguide.com/free/t TCPSlidingWindowAcknowledgmentSystemForDataTranspo.htm)

The TCP Header

The TCP header has 12 fields:

Field	Length	Description
Source Port	16 bits	Source TCP Port
Destination Port	16 bits	Destination TCP Port
Sequence Number	32 bits	Initial Sequence Number
Ack Number	32 bits	Acknowledgement Number
Data Offset	4 bits	Indicates where the data begins in a TCP segment
Reserved	6 bits	Always set to 0
Control Bits	6 bits	URG, ACK, PSH, RST, SYN, and FIN flags
Window	16 bits	Used for Flow Control
Checksum	16 bits	Used for Error-Checking
Urgent Pointer	16 bits	
Options	Variable	
Padding	Variable	To ensure the TCP header ends at a 32 bit boundary

User Datagram Protocol (UDP)

The **User Datagram Protocol (UDP)** is defined as an unreliable, **connectionless** transport protocol. It is essentially a stripped-down version of TCP, and thus has *far less latency* than TCP.

UDP provides no three-way handshake, no flow-control, no sequencing, and no acknowledgment of data receipt. However, UDP does provide basic error-checking using a checksum.

The UDP header has only 4 **fields**:

Field	Length	Description
Source Port	16 bits	Source UDP Port
Destination Port	16 bits	Destination UDP Port
Length	16 bits	Length of the header and the data
Checksum	16 bits	Used for Error-Checking

Comparison of TCP versus UDP

TCP

Connection-oriented
Guaranteed Delivery
Sends Acknowledgments
Reliable, but slow
Segments and Sequences Data
Flow Control
Performs CRC on data

UDP

Connection-less
No Guaranteed Delivery
Does not send Acknowledgments
Unreliable, but fast
Does NOT segment/sequence data
No Flow Control
Performs CRC on data

TCP/UDP Ports

TCP and UDP **ports** identify services that run on a specific logical address. Otherwise, there would be no way to distinguish data destined for one service or another on a device. For example, port numbers allow both a web and email server to operate simultaneously on the same address.

An IP address combined with a TCP or UDP port forms a **socket**. A socket is written out as follows:

10.50.1.1:80

Specific ports (1-1024) have been reserved for specific services, and are recognized as **well-known** ports. Below is a table of several common TCP/UDP ports:

20, 21	TCP	FTP
22	TCP	SSH
23	TCP	Telnet
25	TCP	SMTP
53	UDP	DNS
80	TCP	HTTP
110	TCP	POP3
443	TCP	SSL
666	TCP	Doom

For a complete list of port numbers, refer to the IANA website: http://www.iana.org/assignments/port-numbers.

Section 8 - IPv6 Addressing -

IPv6 Basics

The most widespread implementation of IP currently is IPv4, which utilizes a 32-bit address. Mathematically, a 32-bit address can provide roughly 4 billion unique IP addresses ($2^{32} = 4,294,967,296$). Practically, the number of usable IPv4 addresses is much lower, as many addresses are reserved for diagnostic, experimental, or multicast purposes.

The explosive growth of the Internet and corporate networks quickly led to an IPv4 address shortage. Various solutions were developed to alleviate this shortage, including CIDR, NAT, and Private Addressing. However, these solutions could only serve as temporary fixes.

In response to the address shortage, **IPv6** was developed. IPv6 increases the address size to 128 bits, providing a nearly unlimited supply of addresses (340,282,366,920,938,463,463,374,607,431,768,211,456 to be exact). This provides roughly 50 *octillion* addresses *per person alive* on Earth today, or roughly 3.7×10^{21} addresses per square inch of the Earth's surface.

(References: http://cc.uoregon.edu/cnews/spring2001/whatsipv6.html; http://en.wikipedia.org/wiki/IPv6)

IPv6 offers the following features:

- Increased Address Space and Scalability providing the absurd number of possible addresses stated previously.
- **Simplified Configuration** allows hosts to auto-configure their IPv6 addresses, based on network prefixes advertised by routers.
- **Integrated Security** provides built-in authentication and encryption into the IPv6 network header
- Compatibility with IPv4 simplifies address migration, as IPv6 is backward-compatible with IPv4

The IPv6 Address

The IPv6 address is **128 bits**, as opposed to the 32-bit IPv4 address. Also unlike IPv4, the IPv6 address is represented in hexadecimal notation, separate by colons.

An example of an IPv6 address would be:

1254:1532:26B1:CC14:0123:1111:2222:3333

Each "grouping" (from here on called **fields**) of hexadecimal digits is 16 bits, with a total of eight fields. The hexadecimal values of an IPv6 address are **not case-sensitive.**

We can drop any leading zeros in each field of an IPv6 address. For example, consider the following address:

1423:0021:0C13:CC1E:3142:0001:2222:3333

We can condense that address to: 1423:21:C13:CC1E:3142:1:2222:3333

Only leading zeros can be condensed. If we have an entire field comprised of zeros, we can further compact the following address:

F12F:0000:0000:CC1E:2412:1111:2222:3333

The condensed address would be: F12F::CC1E:2412:1111:2222:3333

Notice the double colons (::). We can only condense one set of contiguous zero fields. Thus, if we had the following address:

F12F:0000:0000:CC1E:2412:0000:0000:3333

We could not condense that to: F12F::CC1E:2412::3333

The address would now be ambiguous, as we wouldn't know how many "0" fields were compacted in each spot. Remember that we can only use one set of double colons in an IPv6 address!

The IPv6 Prefix

IPv4 utilizes a *subnet mask* to define the network "prefix" and "host" portions of an address. This subnet mask can also be represented in Classless Inter-Domain Routing (CIDR) format.

IPv6 always use **CIDR** notation to determine what bits **notate the prefix** of an address:

Full Address: 1254:1532:26B1:CC14:123:1111:2222:3333/64

Prefix ID: 1254:1532:26B1:CC14: *Host ID:* 123:1111:2222:3333

The /64 indicates that the first 64 bits of this address identify the prefix.

The IPv6 Interface ID and EUI-64 Format

The host portion of an IPv4 address is not based on the hardware address of an interface. IPv4 relies on **Address Resolution Protocol (ARP)** to map between the logical IP address and the **48-bit** hardware **MAC address**.

IPv6 unicasts generally allocate the first 64 bits of the address to identify the network (**prefix**), and the last 64 bits to identify the host (referred to as the **interface ID**). The interface ID *is* based on the interface's hardware address.

This interface ID adheres to the IEEE **64-bit Extended Unique Identifier** (**EUI-64**) format. Since most interfaces still use the 48-bit MAC address, the MAC must be converted into the EUI-64 format.

Consider the following MAC address: 1111.2222.3333. The first 24 bits, the Organizationally Unique Identifier (OUI), identify the manufacturer. The last 24 bits uniquely identify the host. To convert this to EUI-64 format:

- 1. The **first 24 bits** of the MAC (the **OUI**), become the first 24 bits of the EUI-64 formatted interface ID.
- 2. The *seventh* bit of the OUI is changed from a "0" to a "1".
- 3. The next 16 bits of the interface ID are **FFFE**.
- 4. The **last 24 bits** of the MAC (the **host ID**), become the last 24 bits of the interface ID.

Thus, the MAC address 1111.2222.3333 in EUI-64 format would become 1311:22FF:FE22:3333, which becomes the interface ID.

The IPv6 Address Hierarchy

IPv4 separated its address space into specific **classes.** The class of an IPv4 address was identified by the high-order bits of the first octet:

- Class A (00000001 011111111, or 1 127)
- Class B (10000000 101111111, or 128 191)
- Class C (11000000 11011111, or 192 223)
- Class D (11100000 11101111, or 224 239)

IPv6's addressing structure is far more scalable. Less than 20% of the IPv6 address space has been designated for use, currently. The potential for growth is enormous.

The address space that *has* been allocated is organized into several types, determined by the high-order bits of the first field:

- **Special Addresses** addresses begin **00***xx*:
- Link Local addresses begin FE8x:
- Site Local addresses begin FECx:
- Aggregate Global addresses begin 2xxx: or 3xxx:
- Multicasts addresses begin FFxx:
- Anycasts

(Note: an "x" indicates the value can be any hexadecimal number)

There are **no broadcast addresses** in IPv6. Thus, any IPv6 address that is not a *multicast* is a *unicast* address.

Anycast addresses identify a group of interfaces on multiple hosts. Thus, multiple hosts are configured with an *identical* address. Packets sent to an anycast address are sent to the *nearest* (i.e., least amount of hops) host. Anycasts are indistinguishable from any other IPv6 unicast address.

Practical applications of anycast addressing are a bit murky. One possible application would be a server farm providing an identical service or function, in which case anycast addressing would allow clients to connect to the nearest server.

<u>Special (Reserved) IPv6 Addresses</u>

The first field of a **reserved** or **special** IPv6 address will always begin **00***xx*. Reserved addresses represent 1/256th of the available IPv6 address space.

Various reserved addresses exist, including:

- 0:0:0:0:0:0:0 (or ::) is an unspecified or unknown address. It is the equivalent of the IPv4 0.0.0.0 address, which indicates the absence of a configured or assigned address. In routing tables, the unspecified address is used to identify all or any possible hosts or networks.
- 0:0:0:0:0:0:1 (or ::1) is the loopback or localhost address. It is the equivalent of the IPv4 127.0.0.1 address.

Reserved Addresses - IPv4 and IPv6 Compatibility

To alleviate the difficulties of immediately migrating from IPv4 to IPv6, specific reserved addresses can be used to *embed* an IPv4 address into an IPv6 address.

Two types of addresses can be used for IPv4 embedding, **IPv4-compatible IPv6 addresses**, and **IPv4-mapped IPv6 addresses**.

• 0:0:0:0:0:0:a.b.c.d (or ::a.b.c.d) – is an IPv4-compatible IPv6 address. This address is used on devices that support both IPv4 and IPv6. A prefix of /96 is used for IPv4-compatible IPv6 addresses:

::192.168.1.1/96

• 0:0:0:0:0:FFFF:a.b.c.d (or ::FFFF:a.b.c.d) – is an IPv6-mapped IPv6 address. This address is used by IPv6 routers and devices to identify *non*-IPv6 capable devices. Again, a prefix of /96 is used for IPv4-mapped IPv6 addresses:

::FFFF:192.168.1.1/96

<u>Link-Local IPv6 Addresses</u>

Link-local IPv6 addresses are used only on a single link (subnet). Any packet that contains a link-local source or destination address is *never routed* to another link. Every IPv6-enabled interface on a host (or router) is assigned a link-local address. This address can be manually assigned, or auto-configured.

The first field of a **link-local** IPv6 address will always begin *FE8x* (1111 1110 10). Link-local addresses are **unicasts**, and represent 1/1024th of the available IPv6 address space. A prefix of /10 is used for link-local addresses.

FE80::1311:22FF:FE22:3333/10

There is no hierarchy to a link-local address:

- The first 10 bits are fixed (FE8), known as the Format Prefix (FP).
- The next 54 bits are set to **0**.
- The final 64 bits are used as the **interface ID**.

Site Local IPv6 Addresses

Site-local IPv6 addresses are the equivalent of "private" IPv4 addresses. Site-local addresses can be routed within a *site* or *organization*, but cannot be globally routed on the Internet. Multiple private subnets within a "site" are allowed.

The first field of a **site-local** IPv6 address will always begin FECx (1111 1110 11). Site-local addresses are **unicasts**, and represent $1/1024^{th}$ of the available IPv6 address space.

FEC0::2731:E2FF:FE96:C283/64

Site-local addresses do adhere to a hierarchy:

- The first 10 bits are the fixed FP (**FEC**).
- The next 38 bits are set to **0**.
- The next 16 bits are used to identify the **private subnet ID**.
- The final 64 bits are used as the **interface ID**.

To identify two separate subnets (1111 and 2222):

FEC0::1111:2731:E2FF:FE96:C283/64 FEC0::2222:97A4:E2FF:FE1C:E2D1/64

<u>Aggregate Global IPv6 Addresses</u>

Aggregate Global IPv6 addresses are the equivalent of "public" IPv4 addresses. Aggregate global addresses can be routed publicly on the Internet. Any device or site that wishes to traverse the Internet must be uniquely identified with an aggregate global address.

Currently, the first field of an **aggregate global** IPv6 address will always begin 2xxx (001). Aggregate global addresses are **unicasts**, and represent $1/8^{th}$ of the available IPv6 address space.

2000::2731:E2FF:FE96:C283/64

Aggregate global addresses adhere to a very strict hierarchy:

- The first 3 bits are the fixed FP.
- The next 13 bits are the **top-level aggregation identifier** (**TLA ID**).
- The next 8 bits are **reserved** for future use.
- The next 24 bits are the **next-level aggregation identifier (NLA ID).**
- The next 16 bits are the **site-level aggregation identifier (SLA ID).**
- The final 64 bits are used as the **interface ID**.

By have multiple **levels**, a consistent, organized, and scalable hierarchy is maintained. High level registries are assigned ranges of TLA IDs. These can then be subdivided in the NLA ID field, and passed on to lower-tiered ISPs.

Such ISPs allocate these prefixes to their customers, which can further subdivide the prefix using the SLA ID field, to create whatever local hierarchy they wish. The 16-bit SLA field provides up to 65535 networks for an organization.

Note: Do not confuse the SLA ID field of a global address field, with a site-local address. Site-local addresses cannot be routed publicly, where as SLA ID's are just a subset of the publicly routable aggregate global address.

Multicast IPv6 Addresses

Multicast IPv6 addresses are the equivalent of IPv4 multicast addresses. Interfaces can belong to one or more multicast **groups**. Interfaces will accept a multicast packet only if they belong to that group. Multicasting provides a much more efficient mechanism than **broadcasting**, which requires that every host on a link accept and process each broadcast packet.

The first field of a **multicast** IPv6 address will always begin **FF***xx* (1111 1111). The full multicast range is **FF00** through **FFFF**. **Multicasts** represent 1/256th of the available IPv6 address space.

FF01:0:0:0:0:0:0:1

Multicast addresses follow a specific format:

- The first 8 bits **identify the address** as a **multicast** (1111 1111)
- The next 4 bits are a **flag value**. If the flag is set to all zeroes (0000), the multicast address is considered *well-known*.
- The next 4 bits are a **scope value**:
 - \circ 0000 (0) = Reserved
 - o 0001 (1) = Node Local Scope
 - o 0010 (2) = Link Local Scope
 - o 0101 (5) = Site Local Scope
 - o 1000 (8) = Organization Local Scope
 - o 1110 (e) = Global Scope
 - \circ 1111 (f) = Reserved
- The final 112 bits identify the actual **multicast group**.

IPv4 multicast addresses had no mechanism to support multiple "**scopes.**" IPv6 scopes allow for a multicast hierarchy, a way to *contain* multicast traffic.

Common IPv6 Multicast Addresses

The following is a list of common, well-known IPv6 multicast addresses:

Node-Local Scope Multicast Addresses

- FF01::1 All-nodes address
- FF01::2 All-routers address

Link-Local Scope Multicast Addresses

- FF02::1 All-nodes address
- FF02::2 All-routers address
- FF02::5 OSPFv3 (OSPF IPv6) All SPF Routers
- FF02::6 OSPFv3 Designated Routers
- FF02::9 RIPng Routers
- FF02::13 PIM Routers

Site-Local Scope Multicast Addresses

• FF05::2 – All-routers address

All hosts must join the **all-nodes** multicast group, for both the node-local and link-local scopes. All routers must join the **all-routers** multicast group, for the node-local, link-local, and site-local scopes.

Every site-local and aggregate global address is assigned a **solicited-node multicast** address. This solicited-node address is created by appending the last 24 bits of the interface ID to the following prefix: FF02::1:FF/103.

Thus, if you have a site-local address of:

FEC0::1111:2731:E2FF:FE96:C283

The corresponding solicited-node multicast address would be:

FF02::1:FF96:C283

Solicited-node multicast addresses are most often used for neighbor discovery (covered in an upcoming section in this guide).

Required IPv6 Addresses

At a minimum, each IPv6 interface on a **host** must recognize the following IPv6 addresses:

- The loopback address
- A link-local address
- Any configured site-local or aggregate global addresses
- Any configured multicast groups
- The all-nodes multicast address (both node-local and link-local scopes)
- The solicited-node multicast address for any configured unicast addresses

In *addition* to the above addresses, each IPv6 interface on a **router** must recognize the following IPv6 addresses:

- The subnet-router anycast address
- Any configured multicast groups
- The all-routers multicast address (node-local, link-local, and site-local scopes)

IPv6 Addresses and URLs

IPv6 addresses can also be referenced in **URLs** (Uniform Resource Locator). URL's, however, use the colon to represent a specific TCP "port". This is not an issue with IPv4 addresses, which can easily be referenced using a URL:

http://192.168.1.1/index.html

Because IPv6 fields are separated by colons, the IPv6 address must be placed in brackets, to conform to the URL standard:

http://[FEC0::CC1E:2412:1111:2222:3333]/index.html

The IPv6 Header

The IPv6 header has **8 fields** and is **320 bits** long. It has been considerably streamlined compared to its IPv4 counterpart, which has **12 fields** and is **160 bits** long.

Field	Length	Description
Version	4 bits	Version of IP (in this case, IPv6)
Traffic Class	8 bits	Classifies traffic for QoS
Flow Label	20 bits	Identifies a flow between a source and destination
Payload Length	16 bits	Length of data in packet
Next Header	8 bits	Specifies the next upper-layer or extension header
Hop Limit	8 bits	Decremented by each router traversed
Source Address	128 bits	Source IPv6 address
Destination Address	128 bits	Destination IPv6 address

The *Next Header* field is of some importance. This field can identify either the next upper-layer header (for example, UDP, TCP or ICMP), or it can identify a special **Extension Header**, which placed in between the IPv6 and upper layer header.

Several such extension headers exist, and are usually processed in the following order:

- **Hop-by-Hop Options** specifies options that should be processed by every router in the path. Directly follows the IPv6 header.
- **Destination Options** specifies options that should be processed by the destination device.
- **Routing Header** specifies each router the packet must traverse to reach the destination (source routing)
- **Fragment Header** used when a packet is larger than the MTU for the path
- **Authentication Header** used to integrate IPSEC Authentication Header (AH) into the IPv6 packet
- **ESP Header** used to integrate IPSEC Encapsulating Security Payload (ESP) into the IPv6 packet

(Reference: http://www.cisco.com/univercd/cc/td/doc/product/software/ios122/122newft/122t/122t2/ipv6/ftipv6o.htm#1004285)

ICMPv6

ICMP Version 6 (**ICMPv6**) is a core component of IPv6. All devices employing IPv6 must also integrate ICMPv6.

ICMPv6 provides many services, including (but not limited to):

- Error Messages
- Informational messages (such as *echo* replies for IPv6 ping)
- MTU Path Discovery
- Neighbor Discovery

There are four key ICMPv6 error messages:

- **Destination Unreachable** (ICMP packet type 1) *indicates that the packet cannot be forwarded to its destination. The node sending this message includes an explanatory code:*
 - o **0** No route to destination
 - o 1 Access is administratively prohibited
 - o 3 Address unreachable
 - o 4 Port unreachable
- Packet Too Big (ICMP packet type 2) indicates the packet is larger than the MTU of the link. IPv6 routers do not fragment packets. Instead, the Packet Too Big message is sent to the source (sending) device, which then reduces (or fragments) the size of the packet to the reported MTU. This message is used for Path MTU Discovery (PMTUD).
- **Time Exceeded** (ICMP packet type 3) *indicates that the hop count limit has been reached, usually indicating a routing loop*
- **Parameter Problem** (ICMP packet type 4) *indicates an error in the IPv6 header, or an IPv6 extension header. The node sending this message includes an explanatory code:*
 - o **0** Erroneous header field
 - 1 Unrecognized next-header type
 - o 2 Unrecognized IPv6 option

(Reference: http://www.cisco.com/en/US/tech/tk365/technologies_tech_note09186a0080113b1c.shtml)

Neighbor Discovery Protocol (NDP) and ICMPv6

The **neighbor discovery protocol** (**NDP**) provides a multitude of services for IPv6 enabled devices, including:

- Automatic address configuration, and prefix discovery
- Duplicate address detection
- MTU discovery
- Router discovery
- Address resolution

NDP replaces many IPv4 specific protocols, such as DHCP and ARP. NDP utilizes **ICMPv6** to provide the above services.

Periodically, IPv6 routers send out **Router Advertisements** (**RA's**) to both announce their presence on a link, and to provide auto-configuration information for hosts. This **RA** (ICMP packet type 134) is sourced from the link-local address of the sending router, and sent to the link-scope all-nodes multicast group. The sending router sets a **hop limit** of **255** on a RA; however, the RA packet *must not* be forwarded outside the local link.

Hosts use RA's to configure themselves, and add the router to its local default router list. A host can request an RA by sending out a **Router Solicitation** (**RS**, ICMP packet type 133) to the link-local all-routers multicast address. A RS is usually sent when a host is *not* currently configured with an IP address.

The RA messages contain the following information for hosts:

- The **router's link-layer address** (to be added to the host's default router list)
- One or more **network prefixes**
- A **lifetime** (measured in seconds) for the prefix(es)
- The link **MTU**

Routers send **Redirect** messages to hosts, indicating a *better* route to a destination. Hosts can have multiple routers in its default router list, but one is chosen as the *true* default router. If this default router deems that another router has a better route to the destination, it forwards the Redirect message to the sending host.

Neighbor Discovery Protocol (NDP) and ICMPv6 (continued)

Neighbor Solicitations (**NS's**, ICMP packet type 135) are sent by hosts to identify the link-layer address of a neighbor, and ensure its reachability. A NS message's source address is the link-local address of the sending host, and the destination is the **solicited-node multicast** address of the destination host.

A neighbor will reply to a NS with a **Neighbor Advertisement** (**NA**, ICMP packet type 136). This process replaces the Address Resolution Protocol (ARP) used by IPv4, and provides a far more efficient means to learn neighbor address information.

Hosts additionally use the NS messages to **detect duplicate addresses**. Before a host assigns itself an IPv6 address, it sends out a NS to ensure no other host is configured with that address.

Autoconfiguration of Hosts

Hosts can be assigned IPv6 addresses one of two ways: manually, or using autoconfiguration. Hosts learn how to autoconfigure themselves from **Router Advertisements (RA's).**

Two types of autoconfiguration exist, stateless and stateful.

When using **Stateless Autoconfiguration**, a host first assigns itself a link-local IPv6 address. It accomplishes this by combining the link-local prefix (FE8) with its interface ID (MAC address in EUI-64 format).

The host then sends a **Router Solicitation** multicast to the all-routers multicast address, which provides one or more network prefixes. The host combines these prefixes with its interface ID to create its site-local (or aggregate global) IPv6 addresses.

Stateful Autoconfiguration is used in conjunction with stateless autoconfiguration. Stateful Autoconfiguration utilizes DHCPv6 to provide additional information to the host, such as DNS servers. DHCPv6 can also be used in the event that there is no router on the link, to provide stateless autoconfiguration.

Section 9 - Introduction to 802.11 Wireless -

802.11 Overview

Beginning in the mid 1990's, the IEEE LAN/MAN committee began developing a series of **Wireless Local Area Network (WLAN)** standards. Collectively, these wireless standards are identified as the **802.11 standard.**

Note: The 802.11 standard is occasionally referred to as **Wi-Fi**, though the term 'Wi-Fi' has been applied to other wireless standards as well.

Various **amendments** have been made to the 802.11 standard. These are identified by the letter appended to the standard, such as 802.11*a* or 802.11*g*. The 802.11 amendments will be covered in greater detail later in this guide.

Wireless devices communicate across a specific range of **RF frequencies** known as a **channel**, using an antenna off of a **radio card**. 802.11 antennas come in several forms:

- Omnidirectional
- Semi-directional
- Highly-directional

A group of communicating 802.11 wireless devices is known as a **service set.** A wireless client can connect point-to-point with another wireless client – this is referred to as an **ad-hoc** connection, or an **Independent Basic Service Set (IBSS).**

More commonly, wireless client are centrally connected via a wireless access point (WAP). This is referred to as an infrastructure connection, or a Basic Service Set (BSS). Wireless clients must associate with a WAP before data can be forwarded. WAPs often serve as a *gateway* between the wired and wireless networks.

In environments where a single WAP does not provide sufficient coverage, multiple WAPs can be *linked* as part of an **Extended Service Set** (**ESS**).

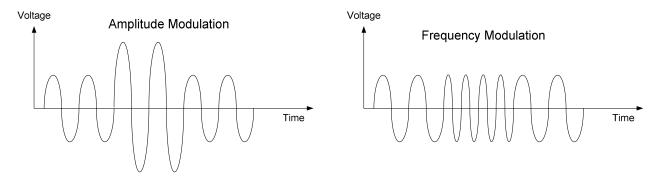
Radio Frequency Overview

Wireless communication is accomplished using **Radio Frequency** (**RF**) waves. **Frequency** is a measurement of the number of cycles completed per a given time period for an electromagnetic wave. The standard frequency measurement unit is the **hertz** (**Hz**), or one *cycle* per *second*.

Note: Ranges of frequencies are often identified by their specific use; these ranges are often referred to as **bands**.

Transmitting devices tune the signal to a specific frequency; receiving devices must tune to this frequency to receive the transmission. A signal at a specific frequency is referred to as the **carrier signal**. However, a carrier signal alone cannot contain data.

Modulation is the method of altering a signal (usually by varying its **amplitude**, **frequency**, or **phase**), to convey a message or data stream:



802.11 devices employ multiple advanced modulation techniques, depending on the 802.11 amendment. This modulation requires that 802.11 devices communicate on a small subset of frequencies (referred to as a **channel**) varying around the carrier signal.

Each 802.11 amendment operates in either the **2.4-GHz** or **5-GHz** band:

- The **2.4-GHz band** provides the greatest range, but is *unregulated* and shared with appliances like microwaves and cordless phones. This can result in interference and degraded performance. The 2.4-GHz band is a subset of the industrial, scientific, and medical (ISM) band.
- The **5-GHz band** is *regulated* and thus generally free of interference. However, signals at this frequency suffer from poor range and are easily obstructed by intermediary objects. The 5-GHz band is referred to as the Unlicensed National Information Infrastructure (UNII) band.

RF Signal Strength

RF signals will attenuate in the open air. The power output of the RF antenna dictates the signal strength, and the usable distance of the signal.

RF power output is not usually measured in absolute terms (such as Watts). Instead, it is measured in **decibels** (**dB**), as a *ratio* of power to a *reference point*. The reference point is usually one Watt (W) or one milliWatt (mW).

The resulting power measurements are **Decibel Watts** (**dBw**) and **Decibel milliWatts** (**dBm**). One dBm is the equivalent of one milliWatt of power output.

Decibel measurements are *logarithmic* in nature. The formula for calculating power output in decibel form is as follows:

$$dB = 10log_{10} (P_{signal} / P_{reference})$$

The abbreviation *P* is short for power. Using the above formula, a signal transmitting at an *absolute* power of 20 mW would be represented as having an *(relative)* output power of 13 dBm.

Because decibels measure a *ratio* of power, it is possible to have a *negative* value. A negative value indicates that the amount of power is *less* than the reference point. For example, .25 mW of absolute power would be represented as -6 dBm. Conversely, a positive value indicates that the amount of power is *more* than the reference point.

RF Interference and Obstruction

In addition to open-air attenuation, RF signals are susceptible to **interference**, degrading the performance and integrity of the communication. As stated previously, other devices operating in the **same frequency range** can interfere with a signal.

Physical objects can also obstruct or alter the trajectory of a RF signal:

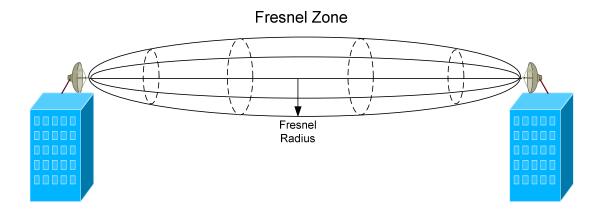
- **Reflection** occurs when a signal *bounces off* of a reflective material, altering its intended trajectory (and sometimes back towards the sender). Metal objects and water often cause reflection. If the signal is reflected in multiple directions, it is referred to as **scattering**.
- **Refraction** occurs when the trajectory of a signal is *bent* as it *passes through* an object, such as a wall.
- **Absorption** occurs when the energy of a signal is *absorbed* as it passes through an object, such as a wall or a tree. This loss of energy degrades the strength of the signal.
- **Diffraction** occurs when a signal *bends around* a signal-absorbing object. For example, a sufficiently-strong signal can bend around an obstructing building, or around a corner within a building. However, this weakens and negatively affects the trajectory of a point-to-point signal.

Because of diffraction, it is particularly important to maintain **line-of-sight** when employing a point-to-point wireless signal over a long distance. Buildings, trees, and even the curvature of the earth can obstruct the line-of-sight of the transmitting/receiving antennas.

(Reference: CCNP BCMSN Official Exam Certification Guide 4th Edition. David Hucaby. Pages 447-450, http://www.cisco.com/warp/public/102/wwan/quick-ref.pdf)

RF Fresnel Zones

Specifically, line-of-sight must be maintained within a signal's elliptical-shaped **Fresnel zone**.



If more than 40% of the lower radius of the Fresnel zone is obstructed, the signal will be negatively impacted from diffraction. Thus, it is imperative to maintain a *minimum* of 60% clearance in this radius.

Calculating the radius of a Fresnel zone requires a complex formula, which is beyond the scope of this guide. Various factors must be accounted for, such as atmospheric refraction, the curvature of the earth, frequency, and the relative heights of the two antennas.

A free Fresnel zone calculator is available online at:

http://www.afar.net/fresnel-zone-calculator/

<u>802.11 Channels</u>

Recall that all amendments to the 802.11 standard operate in one of two frequency bands:

- **2.4-GHz band** (specifically, 2.4000 to 2.4835GHz)
- **5.0-GHz band** (specifically, 5.150 to 5.825GHz)

The 2.4 GHz band supports a **total of 14 channels**, though the FCC limits this to **11 channels** in the United States. The *center* frequency of each channel is separated by only **5 MHz**.

Channel #	Center Frequency	Channel #	Center Frequency
1	2.412 MHz	8	2.447 MHz
2	2.417 MHz	9	2.452 MHz
3	2.422 MHz	10	2.457 MHz
4	2.427 MHz	11	2.462 MHz
5	2.432 MHz	12*	2.467 MHz
6	2.437 MHz	13*	2.472 MHz
7	2.442 MHz	14*	2.484 MHz

^{*} Restricted in US

The 802.11 amendments that use the 2.4-GHz band (specifically, 802.11b and 802.11g) require a **22 MHz range** to modulate the signal. Thus, with each channel's center frequency separated by only 5 MHz, **channel overlap** will occur.

In fact, the 2.4-GHz band supports only **three non-overlapping channels**. Specifically, these are channels **1**, **6**, and **11**. Devices competing on the same or adjacent channels will interfere with each other, degrading performance and reliability.

The less-often used 5-GHz band supports up to **12 non-overlapping channels** (in the U.S.), and is further separated into three sub-bands (with four channels each). The lower and middle bands are dedicated for *indoor* use, and the higher band is dedicated for *outdoor* use.

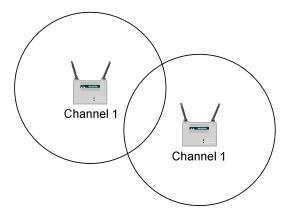
Remember that the 2.4-GHz band is *unregulated*, and the 5.0-GHz band is *regulated*.

(Reference: http://en.wikipedia.org/wiki/List_of_WLAN_channels; http://www.cisco.com/en/US/docs/wireless/access_point/1200/vxworks/configuration/guide/bkscgaxa.html)

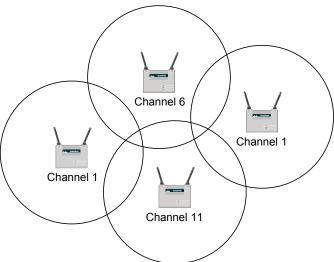
Preventing Channel Overlap

In large environments, a single WAP is often insufficient for full wireless coverage. Multiple WAPs can be linked together as part of an **Extended Service Set (ESS)**.

However, special considerations must be made when installing WAPs in close proximity to each other. Recall that only a limited number of non-overlapping channels are available in both the 2.4-GHz and 5.0-GHz bands.



Adjacent WAPs should *never* be configured on the same channel; the overlapping wireless fields will interfere with each other and *severely* degrade performance.



Providing full wireless coverage while preventing channel overlap can be challenging, especially if the environment has multiple floors. Performing a comprehensive **wireless site survey** is helpful in mapping out an accurate solution.

802.11 and Collisions

If two devices on a half-duplex Ethernet (802.3) network send a packet simultaneously, a **collision** will occur. Similarly, if two 802.11 wireless devices transmit simultaneously, their signals will mix resulting in unusable noise (essentially a *wireless collision*).

Half-duplex 802.3 Ethernet uses **Carrier Sense Multiple Access with Collision Detect (CSMA/CD)** to control media access. Devices monitor the physical link, and will only transmit a frame if the link is idle. When a collision is detected, both devices will wait a random amount of time before resending their respective packets.

All 802.11 connections are half-duplex. The only way to achieve full duplex is to send over one channel, and receive over another. The 802.11 standard currently has no such implementation.

802.11 devices have no method of *detecting* a collision, beyond the failure of the receiving device to send an **acknowledgement**. Instead, 802.11 devices attempt to avoid collisions using **Carrier Sense Multiple Access** with Collision Avoidance (CSMA/CA). Devices will listen before attempting to transmit, and will *only* transmit if no other device is currently transmitting.

If another device *is* transmitting, other devices must wait until that transmission is finished, using a process called **Distributed Coordination Function (DCF)**. The currently transmitting device includes a **duration value** within the 802.11 header, informing other devices of the estimated time-length of its transmission.

Other 802.11 devices will not only wait out this duration value, but will wait an additional random amount of time (referred to as the **DCF interframe space (DIFS)**), before beginning their own transmissions. The random DIFS was implemented to prevent devices from transmitting simultaneously after waiting out another device's transmission duration. DIFS is often referred to as a **random back-off timer**.

(Reference: CCNP BCMSN Official Exam Certification Guide 4th Edition. David Hucaby. Pages 436-438, http://www.cisco.com/en/US/docs/voice_ip_comm/cuipph/7920/5_0/english/design/guide/wrlqos.html#wp1041341)

The 802.11 Amendments

The **original 802.11** standard was release in 1997, and utilized **direct-sequence spread spectrum (DSSS)** to modulate data onto an RF signal. The standard operated in the **2.4 GHz** frequency range, and had a maximum throughput of **2 Mbps**.

The original 802.11 standard never saw widespread adoption, and was quickly supplanted by the 802.11a and 802.11b **amendments**, which were developed concurrently and released in 1999.

802.11 wireless amendments that are currently in deployment include:

• 802.11a

• 802.11g

• 802.11b

• 802.11n

(Reference: http://en.wikipedia.org/wiki/IEEE_802.11_(legacy_mode))

<u>802.11a</u>

The **802.11a** amendment was released in 1999, and utilizes **orthogonal frequency-division multiplexing (OFDM)** for modulation. 802.11a operates in the **5.0-GHz** frequency band, and has a *maximum* throughput of **54 Mbps.** Specifically, 802.11a supports data rates of 6, 9, 12, 18, 24, 36, 48 and 54 Mbps, though the higher throughput is only available in close proximity to the wireless access point (WAP)/transmitter.

Because 802.11a operates in the regulated 5.0-GHz band, it is generally free of interference from other RF devices. However, the higher frequency reduces the effective distance of the signal, and is more susceptible to being absorbed by obstructing objects or walls.

802.11a is generally *not compatible* with other 802.11 amendments, as most of the other amendments operate in the 2.4-GHz band.

In the U.S., 802.11a supports a total of **12** non-overlapping channels, **4** of which can be used outdoors. Despite offering a large number of channels and good throughput, 802.11a did not see the same level of widespread deployment as the less expensive 802.11b and 802.11g amendments.

(Reference: http://en.wikipedia.org/wiki/IEEE_802.11a-1999)

<u>802.11b</u>

The **802.11b** amendment was also released in 1999, and utilizes **complementary code keying (CCK)** for modulation. 802.11b operates in the **2.4-GHz** frequency band, and has a *maximum* throughput of **11 Mbps.** Specifically, 802.11b supports data rates of 1, 2, 5.5, and 11 Mbps.

Because 802.11b operates in the unregulated 2.4-GHz band, it is susceptible to interference from other household RF devices.

In the U.S., 802.11b supports a total of 3 non-overlapping channels, specifically channels 1, 6, and 11.

(Reference: http://en.wikipedia.org/wiki/IEEE 802.11b-1999)

802.11g

The **802.11g** amendment was released in 2003, and utilizes **orthogonal frequency-division multiplexing (OFDM)** for modulation. 802.11g operates in the **2.4-GHz** frequency band, and has a *maximum* throughput of **54 Mbps.** Specifically, 802.11g supports data rates of 6, 9, 12, 18, 24, 36, 48, and 54 Mbps.

As with 802.11b, 802.11g operates in the unregulated 2.4-GHz band, and is susceptible to interference from other household RF devices.

In the U.S., 802.11g supports a total of **3** non-overlapping channels, specifically channels **1**, **6**, and **11**.

802.11g is **backward-compatible** with 802.11b, as they both operate in the 2.4-GHz band. However, if an 802.11b device is present in an 802.11g environment, 802.11g will revert to CCK modulation, and will only support throughputs of 1, 2, 5.5, and 11 Mbps.

Neither 802.11b nor 802.11g are backward-compatible with 802.11a.

(Reference: http://en.wikipedia.org/wiki/IEEE 802.11g-2003)

802.11n

The **802.11n** amendment was officially released in 2009, though pre-release (or *draft*) equipment has been available since 2007. 802.11n supports significantly higher data rates than previous 802.11 amendments, through the use of wider channels (**40MHz** channels instead of 20MHz) and **Multiple-Input Multiple-Output** (**MIMO**).

MIMO employs multiple antennas on both the transmitter and receiver. The resulting multiple data streams are then combined using **Spatial Division Multiplexing (SDM).** This, coupled with 40MHz channels, allows 802.11n to support throughput up to **600 Mbps.**

802.11n devices are identified by the number of *transmit* and *receive* antennas they support, with a format of Antenna_{transmit} x Antenna_{receive}. For example, a WAP with four transmit and three receive antennas would be identified as a 4×3 MIMO WAP.

802.11n can operate in either the **2.4-GHz** *or* the **5.0-GHz** frequency bands, *or* both simultaneously. Thus, 802.11n is backwards compatible with 802.11a, 802.11b, and 802.11g. A pure 802.11n environment should operate in the 5.0-GHz band to maximize throughput and to limit interference.

Note also that the wider 40-MHz channel reduces the number of available non-overlapping channels in each band, which provides more incentive to use the 5.0-GHz band. 802.11n *does* support 20-MHz channels, though this will greatly reduce the maximum throughput.

(Reference: http://en.wikipedia.org/wiki/IEEE_802.11n-2009; http://www.airmagnet.com/assets/whitepaper/WP-802.11nPrimer.pdf; http://www.ciscosystems.sc/en/US/prod/collateral/wireless/ps5678/ps6973/ps8382/prod_white_paper0900aecd806b8ce7_ns767_Networking_Solutions_White_Paper.html)

The 802.11 Amendments - Quick Reference

	<u>802.11</u>	<u>802.11a</u>	802.11b	<u>802.11g</u>	<u>802.11n</u>
Max Throughput	2 Mbps	54Mbps	11Mbps	54Mbps	600Mbps
Modulation	DSSS	OFDM	CCK	OFDM	OFDM
Frequency Band	2.4GHz	5.0GHz	2.4GHz	2.4GHz	2.4/5.0GHz
Non-Overlapping Channels	-	12	3	3	Varies*
Released	1997	1999	1999	2003	2009

^{*} Varies depending on the Frequency Band, and whether 20MHz or 40MHz channels are being utilized.

* * *

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Associating with a Wireless Access Point (WAP)

Recall that a group of communicating 802.11 wireless devices is known as a **service set,** and that there are two *modes* of 802.11 communication:

- Ad-hoc or Independent Basic Service Set (IBSS) where wireless clients communicate point-to-point with each other.
- Infrastructure or Basic Service Set (BSS) where wireless clients communicate via a Wireless Access Point (WAP).

Wireless clients must **associate** with a WAP before data can be forwarded. Various parameters must match between the client and the WAP:

- Service Set Identifier (SSID)
- Data Rate
- Authentication
- Encryption/Data Integrity

The **SSID** is used to *identify* the wireless connection between a WAP (or WAPs) and clients. A wireless client must be configured with the WAP's SSID to associate with it. Otherwise, a client can also *request* (via a *probe*) the SSID if the WAP is configured to **broadcast** the SSID (via a *beacon*). As a best practice, broadcasting is usually disabled in secure environments.

The SSID is often mistaken as a security feature; however, the SSID does not authenticate users or encrypt data – it merely serves as an identifier for a wireless connection. The SSID also provides separation between multiple wireless LANs that might exist in an environment.

Wireless clients are often required to *authenticate* to a WAP. The original 802.11 standard provides for two methods of authentication:

- Open Authentication authenticates any wireless client request.
- **Shared-Key Authentication** requires a matching key to be configured on both the wireless client and WAP.

Open authentication (essentially, *no* authentication) is used for devices that cannot support a complex authentication process. Shared-key authentication employs **Wireless Equivalence Protocol (WEP)** keys for authenticating clients. WEP is covered in detail in the next section.

MAC-address filtering is an additional form of authentication, though not defined in the 802.11 standard. A list of allowed MAC addresses must be maintained on the WAP itself.

Wireless Equivalence Protocol (WEP)

The emergence of 802.11 technologies has introduced new security concerns, due to the open-air nature of wireless transmissions. Such transmissions are easily intercepted, which necessitates mechanisms to not only authenticate wireless clients, but also to secure data transfer (using **encryption**) and to ensure *data integrity* (using a **32-bit CRC**).

Wireless Equivalence Protocol (WEP) was developed as part of the original 802.11 standard. WEP utilizes the **RC4 stream cipher** for encryption, which combines a key with a randomly-generated initialization vector (IV) to provide confidentiality.

WEP comes in two common forms:

- **64-bit WEP** employs a 40-bit key with a 24-bit IV.
- **128-bit WEP –** employs a *104-bit* key with a *24-bit* IV.

The 128-bit WEP key is represented as a 26-digit hexadecimal string.

WEP can be used with both Open and Shared-Key authentication. With Open authentication, the WEP key is used only for encrypting data. With Shared-Key authentication, the WEP key used for both authenticating the wireless client and encrypting data. Regardless of the authentication method, the WEP key(s) **must be identical** on both the wireless client and the WAP.

WEP Shared-Key authentication employs a four-way handshake:

- 1. The client makes an authentication request to the WAP.
- 2. The WAP responds with a clear-text *challenge*.
- 3. The client *encrypts* the challenge with its WEP key.
- 4. The WAP decrypts the encrypted challenge and compares it to the original clear-text challenge.

The authentication process will only be successful if the WEP key is identical on both the WAP and the client. Surprisingly, Shared-Key authentication is less secure than Open authentication. A malicious attacker can intercept both the clear-text and encrypted challenges, and thus somewhat easily derive the encryption key.

WEP is no longer considered a viable security mechanism, as it is *easily* compromised. Additionally, WEP provides only one-way authentication; there is no mechanism within WEP for a client to authenticate the WAP.

(Reference: http://en.wikipedia.org/wiki/Wired_Equivalent_Privacy; http://www.wi-fiplanet.com/tutorials/article.php/1368661)

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Wi-Fi Protected Access (WPA)

Wi-Fi Protected Access (WPA) was developed by the Wi-Fi Alliance to address the shortcomings of WEP. WPA incorporates some of the techniques and protocols that were eventually standardized as part of the 802.11i amendment.

Temporal Key Integrity Protocol (TKIP) is the core component of WPA. Though TKIP employs a **RC4 stream cipher** like WEP, it offers several improvements, including:

- Per-Packet Key Hashing
- 64-bit Message Integrity Check (MIC)
- Broadcast Key Rotation
- Sequence Counting

Note: Cisco developed a proprietary implementation of TKIP that is *not* compatible with WPA TKIP. However, Cisco devices will often support both the standardized and propriety forms of TKIP.

WPA2, also developed by the Wi-Fi Alliance, incorporates *all* portions of the 802.11i amendment. It added support for **Advanced Encryption** Standard (AES) encryption with Cipher Block Chaining Message **Authentication Code Protocol (CCMP)**. AES-CCMP is considered significantly more secure than the RC4 stream cipher used by WEP/TKIP. WPA2 also added native support for Intrusion Detection Systems (IDS).

Both WPA and WPA2 support two *modes*, **Personal** and **Enterprise**.

WPA Personal employs pre-shared key (or passphrase) for authentication, and is often referred to as WPA-PSK (Pre-Shared Key). The WPA key can be represented as a 64-digit hexadecimal string, or an 8 to 63 character ASCII string. As with WEP, this key-string must be identical on both the client and the WAP.

WPA Enterprise employs an 802.1X/EAP server (such as a RADIUS server) for centralized authentication. An authentication server eliminates the need for static encryption/authentication keys to be configured on both the client and the WAP.

802.1X/EAP authentication is covered in detail in the next sections.

(Reference: http://en.wikipedia.org/wiki/Wi-Fi_Protected_Access; http://www.cisco.com/en/US/prod/collateral/wireless/ps5678/ps430/ps4076/prod_brochure09186a00801f7d0b.html)

802.1X and Extensible Authentication Protocol (EAP)

The 802.1X standard was developed by the IEEE to authenticate devices on a Layer-2 port basis. It was originally developed for Ethernet (802.3) bridges and switches, but was expanded to support the authentication of 802.11 wireless devices as well.

802.1X defines three *roles* in the authentication process:

- **Supplicant** the device being authenticated. In an 802.11 environment, the supplicant would be the wireless client software.
- **Authenticator** the device that is *requiring* the authentication. In an 802.11 environment, this is often the WAP.
- **Authentication Server** the device that stores the user database, for validating authentication credentials. This is often an external RADIUS server, though some WAPs support a local user database.

802.1X provides the *encapsulation* of **Extensible Authentication Protocol** (EAP) traffic, which serves as the *framework* for authenticating clients. EAP is not an authentication mechanism in itself. Instead, EAP transports the authentication data between supplicants, authenticators, and authentication servers (all three of which must support 802.1X/EAP).

As a general framework, EAP supports a large number of *methods* for authentication, including (but not limited to):

- Lightweight EAP (LEAP)
- EAP Flexible Authentication via Secure Tunneling (EAP-FAST)
- EAP Transport Layer Security (EAP-TLS)
- Protected EAP (PEAP)

With any form of EAP, wireless clients *must* authenticate with a RADIUS server before any data traffic will be forwarded. Only EAP traffic is allowed between the client and WAP before authentication occurs.

Authenticating clients using 802.1X/EAP offers several advantages over Static-WEP and WPA-PSK, including:

- Centralized management of credentials
- Support for multiple encryption types
- Dynamic encryption keys

 $(Reference: \underline{http://en.wikipedia.org/wiki/IEEE_802.1X; \ \underline{http://en.wikipedia.org/wiki/Extensible_Authentication_Protocol; \ \underline{http://www.ieee802.org/1/files/public/docs2000/P8021XOverview.PDF})$

Lightweight Extensible Authentication Protocol (LEAP)

Lightweight Extensible Authentication Protocol (LEAP) was developed by Cisco, and is supported by WPA/WPA2 as an 802.1X authentication method. LEAP employs a username/password for authentication via a RADIUS server, and does *not* require the use of certificates.

LEAP is supported by most operating system, including Mac OS, Linux, DOS, and most versions of Windows. LEAP additionally supports single sign-on in Windows environments, allowing clients to perform Active Directory (or NT Domain) and 802.1X authentication simultaneously.

LEAP authentication is a multi-step process:

- 1. The supplicant initiates the connection with a *Start* message.
- 2. The authenticator responds with a *Request/Identity* message.
- 3. The supplicant responds with an *Identity* message containing a username.
- 4. The authenticator then forwards the username to the authentication server with an Access Request message.
- 5. The supplicant and authentication server then authenticate each other using a challenge/response method. The authentication server sends a randomly-generated *challenge* to the supplicant. The supplicant then generates a hash value from the challenge and its password, using MD5. This hash value serves as the *response* back to the authentication server, and eliminates the need for the actual password to be transmitted between the two devices.
- 6. A Success message is generated if the supplicant and authentication server have successfully authenticated each other, which informs the authenticator that the supplicant can now pass data traffic.

Once authentication is completed, the supplicant and authentication server then generate a pairwise master key (PMK). The PMK is used to create the actual encryption keys for data transfer, via a four-way handshake.

LEAP was built on a variation of MS-CHAP, and is thus vulnerable to dictionary attacks. A strong password policy is extremely important when employing LEAP in a business environment. If strong passwords are not possible, Cisco recommends utilizing EAP-FAST instead of LEAP.

(Reference: http://www.ciscosistemi.com/en/US/prod/collateral/wireless/ps5678/ps430/prod_qas0900aecd801764f1.html; http://www.cisco.com/application/pdf/en/us/guest/netsol/ns386/c649/ccmigration_09186a0080871da5.pdf; CCNP ONT Exam Certification Guide, Amir Ranjbar. Pages 262-264)

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EAP with Flexible Authentication via Secure Tunneling (EAP-FAST)

EAP-FAST was also developed by Cisco as an alternative to LEAP, and was standardized by the IETF. Like LEAP, it utilizes a username/password for authentication via a RADIUS server, and does not require the use of certificates. Unlike LEAP, EAP-FAST is not vulnerable to dictionary attacks, as it establishes a secure tunnel between the supplicant and authentication server.

EAP-FAST is supported by most versions of Windows, and supports Windows single sign-on in Active Directory/Domain environments.

EAP-FAST authentication is a three-phase process:

- Phase 0 (optional) the supplicant is assigned a Protected Access Credential (PAC), on a per-user basis. This phase is optional because the PAC can be manually configured on the supplicant.
- Phase 1 the supplicant and authentication server establish a secure tunnel using the PAC.
- Phase 2 the supplicant sends its username/password credentials to the authentication server, via the secure tunnel.

(Reference: http://www.cisco.com/en/US/prod/collateral/wireless/ps5679/ps5861/prod_qas09186a00802030dc_ps430_Products_Q_and_A_Item.html; CCNP ONT Exam Certification Guide, Amir Ranjbar. Pages 264-266)

EAP with Transport Layer Security (EAP-TLS)

EAP with **Transport Layer Security (EAP-TLS)** is an IETF standard protocol, and was the *first* EAP authentication method used with 802.11 wireless networks.

EAP-TLS utilizes Public Key Infrastructure (PKI) to authenticate supplicants using certificates. Both the supplicant and the authentication server must be assigned a certificate from a Certificate Authority (CA) server. Because of this, EAP-TLS is considered *extremely* secure, though the complexity of client-side certificates makes it somewhat unpopular.

EAP-TLS is natively supported on most versions of Windows (2000 and newer).

(Reference: http://www.cisco.com/en/US/tech/tk722/tk809/technologies_white_paper09186a008009256b.shtml; CCNP ONT Exam Certification Guide, Amir Ranjbar. Pages 266-267)

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Protected EAP (PEAP)

Protected EAP (PEAP) was developed jointly by Cisco, Microsoft, and RSA Security, and was submitted to the IETF for standardization.

PEAP utilizes TLS to create a secure tunnel between the supplicant and authentication server. The key difference between PEAP and EAP-TLS is that only the *authentication server* requires a PKI certificate – no certificate is required on the supplicant.

PEAP authentication is a two-phase process:

- 1. The supplicant authenticates the authentication server by verifying the server-side PKI certificate. If successful, the supplicant and authentication server form the TLS tunnel.
- 2. The supplicant sends its username/password credentials to the authentication server, via the secure tunnel. This is accomplished using either EAP-MSCHAPv2 (for Windows-based authentication servers) or EAP-GTC (Generic Token Card, for LDAP-based authentication servers).

As with the other EAP-methods, a *Success* message is generated if the supplicant and authentication server have successfully authenticated each other, which informs the authenticator that the supplicant can pass traffic. Part II

The Cisco IOS

Section 10 - Router Components -

Router Memory Components

Cisco routers (and switches) generally contain four types of memory:

- ROM (Read-Only Memory)
- Flash
- NVRAM (Non-Volatile RAM)
- RAM (Random-Access Memory)

ROM contains a **bootstrap** program called **ROM Monitor** (or **ROMmon**). When a router is powered on, the bootstrap runs a hardware diagnostic called **POST** (**Power-On Self Test**).

If POST completes successfully, the bootstrap then attempts to locate and load the Cisco IOS (Internetwork Operating System) stored in Flash **memory.** Flash memory can be erased or overwritten, thus making the Cisco IOS upgradeable. The Cisco IOS is covered in great detail in other guides.

If the bootstrap *cannot* find the IOS in Flash, a stripped-down version of the IOS that will be loaded from ROM instead. The contents of ROM cannot be altered or erased; the entire ROM chip must be replaced if an upgrade/repair is necessary.

If the bootstrap does find the IOS in Flash, it is loaded into RAM and attempts to find a **Startup Configuration** (startup-config) file in **NVRAM**. NVRAM is non-volatile, thus its contents will survive a power-cycle.

If the IOS *cannot* find a startup-config file in NVRAM, it will attempt to load a configuration file from a **TFTP** server (this request is **broadcasted** to 255.255.255). If no TFTP server responds, the IOS will enter Initial **Configuration Mode**, a series of interactive questions intended for quick configuration of the router.

If the IOS does find a startup-config file in NVRAM, this file is loaded into **RAM**, and becomes the **Running Configuration** (running-config). RAM is a volatile memory, and thus its contents will be lost if the router is powercycled.

Router Memory, Quick Reference

The following table details each of the basic types of **router memory**:

<u>Memory</u>	Writable?	<u>Volatile?</u>	<u>Function</u>
ROM	No	No	Stores bootstrap
Flash	Yes	No	Stores IOS
NVRAM	Yes	No	Stores startup-config
RAM	Yes	Yes	Stores running-config

The Router Boot-Process, Quick Reference

The following details the **router boot process**:

- 1. The router is powered on.
- 2. The bootstrap program (ROMmon) is loaded from ROM.
- 3. The bootstrap runs POST.
- 4. The bootstrap attempts to load the IOS from Flash.
 - a. If the IOS *is not* found in Flash, the bootstrap loads into RAM the basic IOS stored in ROM.
 - b. If the IOS is found in Flash, it is loaded into RAM.
- 5. The IOS attempts to load the startup-config file from NVRAM
 - a. If the startup-config is not found in NVRAM, the IOS attempts to load a configuration file from TFTP.
 - b. If no TFTP server responds, the router enters Initial Configuration Mode.
 - c. If the startup-config is found in NVRAM, it is loaded into RAM.
- 6. The startup-config becomes the running-config in RAM.

Interfaces vs Lines

Cisco devices contain two distinctly different types of ports, **interfaces** and lines.

Interfaces connect routers and switches to each other. In other words, traffic is actually routed or switched across interfaces. Examples of interfaces include (but are not limited to):

- Serial interfaces
- Ethernet interfaces
- Fast Ethernet interfaces
- Token Ring interfaces
- ATM interfaces

Interfaces are identified by both the type of interface, and the **interface** number (which always begins at "0"). Thus, the first Ethernet interface on a router would be identified as *Ethernet0*.

Certain router families (such as the 3600 series) are modular, and have multiple "slots" for interfaces. Thus, interfaces on these routers are identified by both the **module number** and the interface number, formatted as: module/interface. Thus, the third Fast Ethernet interface on the first modular slot would be identified as FastEthernet0/2.

Lines identify ports that allow us to connect into, and then configure, Cisco devices. The most common examples of lines include:

- Console ports
- Auxiliary ports
- VTY (telnet) ports

Just like interfaces, lines are identified by both the type of line, and the line number (again, always begins at "0"). Thus, the first console port on a router would be identified as Console0.

Using Lines to Configure the IOS

As mentioned previously, three methods (or *lines*) exist to configure Cisco IOS devices:

- Console ports
- Auxiliary ports
- VTY (telnet) ports

Nearly every modern Cisco router or switch includes a **console port**, sometimes labeled on the device simply as con. The console port is generally a RJ-45 connector, and requires a **rollover** cable to connect to. The opposite side of the rollover cable connects to a PC's serial port using a serial terminal adapter.

From the PC, software such as HyperTerminal is required to make a connection from the local serial port to the router console port. The following settings are necessary for a successful connection:

- Bits per second 9600 baud
- Data bits 8
- Parity None
- Stop bits 1
- Flow Control Hardware

Some Cisco devices include an **auxiliary port**, in addition to the console port. The auxiliary port can function similarly to a console port, and can be accessed using a rollover cable. Additionally, auxiliary ports support modem commands, thus providing dial-in access to Cisco devices.

Telnet, and now SSH, are the most common methods of remote access to routers and switches. The standard edition of the IOS supports up to 5 simultaneous VTY connections. Enterprise editions of the IOS support up to 255 VTY connections.

There are two requirements before a router/switch will accept a VTY connection:

- An **IP** address must be configured on an interface
- At least one VTY port must be configured with a **password**

Section 11 - Introduction to the Cisco IOS -

Cisco IOS

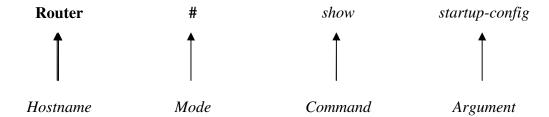
The Cisco IOS (Internetwork Operating System) is a command-line interface used by nearly all current Cisco routers and Catalyst switches. The IOS provides the mechanism to configure all Layer 2 and Layer 3 functions on Cisco devices.

The IOS is structured into several **modes**, which contain sets of commands specific to the function of that mode. Access to a specific mode (and specific commands) is governed by **privilege levels**. (Both modes and privilege levels are covered in great detail in this guide).

The following is a representation of the IOS command-line interface, with an example command:

Router# show startup-config

All commands throughout all guides on this site will be represented like the above. The following is an explanation of each component of the above command:



Hitting the "enter" key after a command will usually yield output specific to your command:

```
Router# show startup-config
version 12.2
service timestamps log uptime
service password-encryption
hostname Router
<snip>
```

(Note: The above output was truncated to save space.)

IOS Version Numbers

IOS version numbers are formatted as follows:

x.y(z)t

- The "x" designates a major revision number.
- The "y" designates a minor revision number.
- The "z" designates an individual release number
- The "t" designates a train identifier.

Thus, the third release of IOS version 12.4 would be identified as 12.4(3). The major and minor revision numbers combined is often called the Maintenance Release number (for example, "12.4").

Trains identify IOS releases to specific markets, and are represented by a single letter:

- The "T" or **Technology** train is continuously updated with new features and security fixes.
- The "E" or **Enterprise** train contains features and commands for enterprise-level equipment.
- The "S" or **Service Provider** train contains features and a command-set for specific ISP equipment

The absence of a train identifier denotes a **Mainline** release. Security updates are released for the mainline train, but new functionality is never added to the feature set.

The latest version of the IOS (as of this writing) is 12.4(11)T. To view the IOS version of your Cisco device:

Router# show version

The Cisco IOS is stored in **Flash** on Cisco routers and Catalyst switches, in a .BIN file format. It can be upgraded using one of several methods:

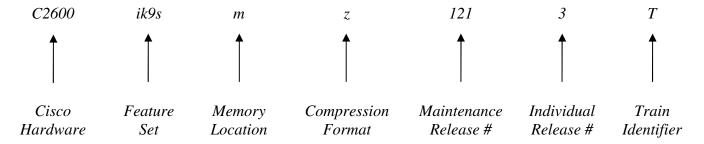
- Replacing the existing Flash stick
- Via a TFTP server
- Via Xmodem
- Via a PCMCIA slot (not supported by all Cisco devices)

(Reference: http://en.wikipedia.org/wiki/Cisco_IOS)

IOS Version Numbers (continued)

The IOS .bin file stored in flash follows a specific naming convention. Observe the following IOS image:

The following is an explanation of each component of the above file name:



(Reference: http://www.cisco.com/en/US/products/sw/iosswrel/ps1828/products_white_paper09186a008018305e.shtml)

The IOS supports a wide variety of feature sets. The following is a list of common feature sets (and is by no means comprehensive):

- ipbase
- ipvoice
- advsecurityk9
- advipservicesk9
- ik9s
- jk9s
- io3
- bin

(Reference: http://www.cisco.com/en/US/products/hw/routers/ps259/prod_bulletin09186a0080161082.html)

IOS Modes on Cisco Devices

As stated earlier in this guide, the Cisco IOS is comprised of several modes, each of which contains a set of commands specific to the function of that mode.

By default, the first mode you enter when logging into a Cisco device is **User EXEC mode.** User mode appends a ">" after the device hostname:

Router>

No configuration can be changed or viewed from User mode. Only basic status information can be viewed from this mode.

Privileged EXEC mode allows all configuration files, settings, and status information to be viewed. Privileged mode appends a "#" after the device hostname:

Router#

To enter Privileged mode, type *enable* from User mode:

Router> *enable* Router#

To return back to User mode from Privileged mode, type *disable*:

Router# disable **Router>**

Very little configuration can be *changed* directly from Privileged mode. Instead, to actually configure the Cisco device, one must enter Global **Configuration mode:**

Router(config)#

To enter Global Configuration mode, type *configure terminal* from Privileged Mode:

> Router# configure terminal Router(config)#

To return back to Privileged mode, type *exit*:

Router(config)# exit Router#

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IOS Modes on Cisco Devices (continued)

As its name implies, Global Configuration mode allows parameters that globally affect the device to be changed. Additionally, Global Configuration mode is sectioned into several sub-modes dedicated for specific functions. Among the most common sub-modes are the following:

- Interface Configuration mode Router(config-if)#
- Line Configuration mode -Router(config-line)#
- Router Configuration mode -Router(config-router)#

Recall the difference between *interfaces* and *lines*. **Interfaces** connect routers and switches to each other. In other words, traffic is actually routed or switched across interfaces. Examples of interfaces include Serial, ATM, Ethernet, Fast Ethernet, and Token Ring.

To configure an interface, one must specify both the type of interface, and the interface *number* (which always begins at "0"). Thus, to configure the first Ethernet interface on a router:

> **Router(config)**# interface ethernet 0 Router(config-if)#

Lines identify ports that allow us to connect into, and then configure, Cisco devices. Examples would include console ports, auxiliary ports, and VTY (or telnet) ports.

Just like interfaces, to configure a line, one must specify both the *type* of line, and the line *number* (again, always begins at "0"). Thus, to configure the first console line on a router:

> **Router(config)**# line console 0 Router(config-line)#

Multiple telnet lines can be configured simultaneously. To configure the first five telnet (or VTY) lines on a router:

> Router(config)# line vty 0 4 Router(config-line)#

Remember that the numbering for both interfaces and lines begins with "0."

Router Configuration mode is used to configure dynamic routing protocols, such as RIP. This mode is covered in great detail in other guides.

IOS Command Shortcuts

Shortcuts are allowed on the IOS command-line, as long as the truncated command is not ambiguous. For example, observe the following commands:

> **Router**# clear Router# clock Router# configure Router# connect Router# copy Router# debug

We could use de as a shortcut for the debug command, as no other command here begins with de. We could not, however, use co as a shortcut, as three commands begin with those letters. The following error would be displayed:

Router# co

% Ambiguous command: "co"

If you type a command incorrectly, the IOS will point out your error:

Router# clcok

% Invalid input detected at "^" marker

Keyboard Shortcuts

Several hotkeys exist to simplify using the IOS interface:

Keyboard Shortcut

Result

CTRL-B (or Left-Arrow)	Moves cursor back one character
CTRL-F (or Right-Arrow)	Moves cursor forward one character
CTRL-A	Moves cursor to beginning of a line
CTRL-E	Moves cursor to end of a line
ESC-B	Moves cursor back one word
ESC-F	Moves cursor forward one word
CTRL-P (or Up-Arrow)	Returns previous command(s) from history buffer
CTRL-N (or Down-Arrow)	Returns next command from history buffer
CTRL-Z	Exits out of the current mode
TAB	Finishes an incomplete command (assuming it is
	not ambiguous)

(Reference: http://www.cisco.com/en/US/products/hw/switches/ps708/products_configuration_guide_chapter09186a008007e6d5.html#wp1028871)

* * *

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Terminal History Buffer

As implied in the previous section, the Cisco IOS keeps a history of previously entered commands. By default, this history buffer stores the previous 10 commands entered. To view the terminal history buffer:

RouterA# *show history*

enable config t hostname RouterA exit show history

The **Up-Arrow** key (or **CTRL-P**) allows you to scroll through previously entered commands. To scroll back down the list, use the **Down-Arrow** key (or CTRL-N).

To adjust the number of commands the history buffer stores (range 0-256):

RouterA# terminal history size 30

IOS Context-Sensitive Help

The **question mark** (?) is one of the most powerful tools in the Cisco IOS, as it provides **context-sensitive help** for each IOS mode.

Typing? at the command prompt displays a list of all commands available at that mode, with explanations:

Router# ?

```
access-enable
           Create a temporary Access-List entry
access-profile Apply user-profile to interface
access-template Create a temporary Access-List entry
     ALPS exec commands
<snip>
```

Typing in part of a command with a ? displays a list of all commands that begin with those characters:

```
Router# co?
configure connect copy
```

Typing in a full command followed by a ? displays the available options and arguments for that command:

```
Router# clock?
set
     Set the time and date
```

Notice the space between the command *clock* and the ?.

The "Show" Command

The *show* command provides the ability to view a wide variety of configuration and status information on your router. The command is executed from Privileged mode, and the syntax is simple:

```
Router# show [argument]
```

There are literally dozens of arguments for the *show* command, and each provides information on a specific aspect of the router. Numerous show commands will be described throughout this and most other guides.

One common *show* command displays the IOS version, configurationregister settings, router uptime, and basic router hardware information:

Router# show version

```
Cisco Internetwork Operating System Software
IOS (tm) 2500 Software (C2500-IS-L), Version 12.3(1a), RELEASE SOFTWARE (fc1)
Copyright (c) 1986-2003 by cisco Systems, Inc.
Compiled Fri 09-Jan-03 11:23 by xxxxx
Image text-base: 0x0307F6E8, data-base: 0x00001000
ROM: System Bootstrap, Version 11.0(10c)XB2, PLATFORM SPECIFIC RELEASE SOFTWARE
(fc1)
BOOTLDR: 3000 Bootstrap Software (IGS-BOOT-R), Version 11.0(10c)XB2, PLATFORM
SPECIFIC RELEASE SOFTWARE (fc1)
Router uptime is 2 minutes
System returned to ROM by reload
System image file is "flash:c2500-is-1.123-1a.bin"
cisco 2500 (68030) processor (revision L) with 14336K/2048K bytes of memory.
Processor board ID 13587050, with hardware revision 00000000
Bridging software.
X.25 software, Version 3.0.0.
2 Ethernet/IEEE 802.3 interface(s)
2 Serial network interface(s)
32K bytes of non-volatile configuration memory.
16384K bytes of processor board System flash (Read ONLY)
Configuration register is 0x2102
```

The following command provides output similar to *show version*:

Router# show hardware

Other common *show* commands will be described shortly.

(Example show version output from: http://www.cisco.com/en/US/products/hw/routers/ps233/products_tech_note09186a008009464c.shtml)

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<u>Enable Passwords</u>

The *enable* password protects a router's Privileged mode. This password can be set or changed from Global Configuration mode:

> **Router(config)**# enable password MYPASSWORD Router(config)# enable secret MYPASSWORD2

The *enable password* command sets an unencrypted password intended for legacy systems that do not support encryption. It is no longer widely used.

The *enable secret* command sets an MD5-hashed password, and thus is far more secure. The *enable password* and *enable secret* passwords **cannot be identical**. The router will not accept identical passwords for these two commands.

Line Passwords and Configuration

Passwords can additionally be configured on router **lines**, such as telnet (vty), console, and auxiliary ports. To change the password for a console port and all telnet ports:

Router(config)# line console 0 **Router(config)#** *line vty 0 4* Router(config-line)# login Router(config-line)# login

Router(config-line)# password cisco1234 Router(config-line)# password cisco1234

Router(config-line)# exec-timeout 0 0 **Router(config-line)**# exec-timeout 0 0 **Router(config-line)**# logging synchronous **Router**(config-line)# logging synchronous

The exec-timeout 0 0 command is optional, and disables the automatic timeout of your connection. The two zeroes represent the timeout value in minutes and seconds, respectively. Thus, to set a timeout for 2 minutes and 30 seconds:

Router(config-line)# exec-timeout 2 30

The *logging synchronous* command is also optional, and prevents system messages from interrupting your command prompt.

By default, line passwords are stored in clear-text in configuration files. To ensure these passwords are encrypted in all configuration files:

Router(config)# service password–encryption

Router Interfaces

Recall that, to configure an interface, one must specify both the type of interface, and the interface *number* (which always begins at "0"). Thus, to configure the first Ethernet interface on a router:

> **Router(config)**# interface ethernet 0 Router(config-if)#

Certain router families (such as the 3600 series) are modular, and have multiple "slots" for interfaces. All commands must reflect both the **module number** and the interface number, formatted as: module/interface

Thus, to configure the third Fast Ethernet interface off of the first module:

Router(config)# interface fastethernet 0/2 Router(config-if)#

By default, all router interfaces are **administratively shutdown**. To take an interface out of an administratively shutdown state:

> **Router(config)#** interface fa 0/0 Router(config-if)# no shutdown

Notice the use of fa as a shortcut for fastethernet in the above example. To manually force an interface into a shutdown state:

Router(config-if)# shutdown

To assign an IP address to an interface:

Router(config-if)# ip address 192.168.1.1 255.255.255.0

An additional *secondary* IP Address can be assigned to an interface:

Router(config-if)# ip address 192.168.1.1 255.255.255.0 **Router(config-if)**# *ip address* 192.168.1.2 255.255.255.0 secondary

Serial interfaces require special consideration. The DCE (Data **Communication Equipment)** side of a serial connection must set the speed, or clock rate, for the DTE (Data Terminal Equipment) side. Clock rate is measured in BPS (bits-per-second).

To set the *clock rate*, if you are the DCE side of a serial connection:

Router(config)# *interface serial 0* Router(config-if)# clock rate 64000

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Status of Router Interfaces

To view the current status and configuration of all interfaces:

Router# show interfaces

```
Ethernet 0 is up, line protocol is up
 Hardware is Ethernet, address is 5520.abcd.1111
 Internet address is 192.168.1.1, subnet mask is 255.255.255.0
 MTU 1500 bytes, BW 10000 Kbit, DLY 100000 usec, rely 255/255, load 1/255
 Encapsulation ARPA, loopback not set, keepalive set (10 sec)
 ARP type: ARPA, ARP Timeout 4:00:00
  Last input 0:00:00, output 0:00:00, output hang never
  Last clearing of "show interface" counters 0:00:00
  Output queue 0/40, 0 drops; input queue 0/75, 0 drops
 Five minute input rate 0 bits/sec, 0 packets/sec
 Five minute output rate 2000 bits/sec, 4 packets/sec 53352 packets input, 351251 bytes, 0 no buffer
     Received 4125 broadcasts, 0 runts, 0 giants, 0 throttles
     0 input errors, 0 CRC, 0 frame, 0 overrun, 0 ignored, 0 abort
     12142 packets output, 16039 bytes, 0 underruns
     O output errors, O collisions, O interface resets, O restarts
Ethernet 1 is up, line protocol is up
  Hardware is Ethernet, address is 5520.abcd.1112
  <snip>
```

The *show interfaces* command displays a plethora of information, including:

- Current interface status (ethernet 0 is up, line protocol is up)
- MAC address (5520.abcd.1111)
- IP address (192.168.1.1)
- MTU (1500 bytes)
- Bandwidth (10 Mbps)
- Output and input queue status
- Traffic statistics (packets input, packets output, collisions, etc.)

To view the current status of a *specific* interface:

Router# show interfaces ethernet 0

To view only IP information for all interfaces:

Router# *show ip interface brief*

Interface	IP Address	OK?	Method	Status	Protocol
Ethernet0 Ethernet1 Serial0	192.168.1.1 192.168.2.1 unassigned	YES	NVRAM	±	up up down

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Status of Router Interfaces (continued)

Traffic can only be routed across an interface if its status is as follows:

Serial 0 is up, line protocol is up

The first part of this status (SerialO is up) refers to the **physical layer** status of the interface. The second part (line protocol is up) refers to the data-link **layer** status of the interface. A status of *up/up* indicates that the physical interface is active, and both sending and receiving keepalives.

An interface that is physically down will display the following status:

Serial 0 is down, line protocol is down

The mostly likely cause of the above status is a defective (or unplugged) cable or interface.

There are several potential causes of the following status:

Serial 0 is up, line protocol is down

Recall that *line protocol* refers to data-link layer functions. Potential causes of the above status could include:

- Absence of keepalives being sent or received
- Clock rate not set on the DCE side of a serial connection
- Different encapsulation types set on either side of the link

An interface that has been administratively shutdown will display the following status:

Serial 0 is administratively down, line protocol is down

Managing Configuration Files

Cisco IOS devices employ two distinct configuration files

- running-config stored in RAM, contains the *active* configuration
- **startup-config** stored in NVRAM (or flash), contains the *saved* configuration

Any configuration change made to an IOS device is made to the runningconfig. Because the running-config file is stored in RAM, the contents of this file will be lost during a power-cycle. Thus, we must save the contents of the running-config to the startup-config file. We accomplish this by using the *copy* command from Privileged mode:

Router# copy running-config startup-config

The copy command follows a very specific logic: copy [from] [to]. Thus, if we wanted to copy the contents of the startup-config file to running-config:

Router# copy startup-config running-config

We can use shortcuts to simplify the above commands:

Router# copy run start **Router#** copy start run

To view the contents of the running-config and startup-config files:

Router# show run **Router#** show start

To delete the contents of the startup-config file:

Router# erase start

If the router is power-cycled after erasing the startup-config file, the router will enter Initial Configuration Mode (sometimes called Setup Mode). This mode is a series of interactive questions intended for quick reconfiguration of the router.

Initial Configuration Mode can be exited by typing CTRL-C.

"Piping" Commands

In newer versions of the Cisco IOS, the output of show commands can be filtered to remove irrelevant lines, through the use of the **pipe** "|" character.

The following command will display the contents of the startup-config, beginning with the first line containing the text ethernet:

Router# show startup | begin ethernet

The following command will **exclude** all lines containing the text *ethernet*:

Router# *show startup* | *exclude ethernet*

The following command will **include** all lines containing the text *ethernet*:

Router# *show startup* | *include ethernet*

Miscellaneous Commands

To change the hostname of your router:

Router(config)# hostname MyRouter **MyRouter**(**config**)# hostname MyRouter

To assign a description to an interface for documentation purposes:

Router(config)# interface serial 0 Router(config-if)# description SBC T1 connection to Chicago

Router# show interfaces

```
Serial 0 is up, line protocol is up
 Hardware is Serial
  Internet address is 70.22.3.1, subnet mask is 255.255.255.0
 Description: SBC T1 connection to Chicago
```

To create a *banner* message which users will see when logging into an IOS device:

Router(config)# banner motd #

```
Logging into this router without authorization is illegal
and will be prosecuted!
```

The # symbol is used as a delimiter to indicate the beginning and end of the banner. Any character can be used as a delimiter.

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IOS Troubleshooting Commands

The show tech-support command prints to screen every configuration file, and the output of several important show commands. This can be redirected to a file and either viewed or sent to Cisco for troubleshooting purposes:

Router# *show tech-support*

The *debug* command is a powerful tool to view real-time information and events occurring on an IOS device. As with the *show* command, there are a multitude of arguments for the *debug* command. An example *debug* command is as follows:

Router# debug ip rip events

To disable a specific debugging command, simply preprend the word no in front of the command:

Router# *no debug ip rip events*

To enable all possible debugging options on an IOS device:

Router# debug all

Using the *debug all* command is not recommended, as it will critically impair router performance.

To disable all possible debugging options on an IOS device:

Router# no debug all

Section 12 - Advanced IOS Functions -

The Configuration Register

The **configuration register** (**config-register**) is a hexadecimal value that controls various aspects of how a router boots, including:

- Baud Rate
- Boot Messages (enable/disable)
- Break (disable/ignore)
- Flash (read-only, read-write)
- NVRAM (use startup-config/bypass startup-config)

The **default** config-register is **0x2102.** To view your router's current configregister setting:

Router# show version

```
<snip>
32K bytes of non-volatile configuration memory.
16384K bytes of processor board System flash (Read ONLY)
Configuration register is 0x2102
```

Common config-register settings include:

<u>Value</u>	Baud Rate	Boots Into?	<u>Flash</u>	Startup-Config
0x2101	9600	IOS from ROM	Read/Write	Uses
0x2102	9600	IOS from Flash	Read/Only	Uses
0x2142	9600	IOS from Flash	Read/Only	Bypass

Remember, 0x2102 is the default config-register value on Cisco routers. In order to upgrade the Cisco IOS, the config-register must be changed to **0x2101**, so that the Flash memory becomes writeable.

To change the config-register from the IOS:

Router(config)# config-register 0x2142

This configuration change **does not take affect** until the **next reboot**.

(Reference: http://www.cisco.com/en/US/products/hw/routers/ps133/products_tech_note09186a008022493f.shtml)

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Backing up and Restoring the Cisco IOS

The Cisco IOS is stored in flash. Multiple IOS files can be loaded into flash, assuming there is enough free space. You can view available free space, and the name of any file(s) in flash, by typing:

Router# show flash

```
System flash directory:
File Length Name/status
      4467254 c2500-ik9s-mz.122-4
[4467254 bytes used, 12309962 available, 16777216 total]
16384K bytes of processor board System flash (Read/Write)
```

To backup the IOS, a TFTP server is required. The TFTP server must have IP connectivity to the router.

To backup the IOS file from the router's flash to the TFTP server:

```
Router# copy flash tftp
```

You will be prompted for the following information:

- Address of remote host (the TFTP server)
- Source file name (the name of the file in flash)
- Destination file name

To load an IOS file from a TFTP server to the router's flash:

```
Router# copy tftp flash
```

The process is nearly identical to copy a startup-configuration file to or from a router's NVRAM:

> **Router**# *copy startup tftp* Router# copy tftp startup

CDP (Cisco Discovery Protocol)

CDP is a Cisco propriety protocol used to collect information about locally attached Cisco switches and routers. CDP is enabled by default on all IOS enabled routers and switches, and sends out updates every 60 seconds.

CDP will provide the following information about directly connected neighbors:

- **Device ID** hostname of remote router/switch
- Local Interface interface that remote router/switch is directly connected to
- **Holdtime** amount of time before remote device information is purged from CDP table
- Capability Type of remote device (router, switch, host)
- **Platform** Model of remote device

To view CDP timers and hold-down information:

Router# show cdp

To display neighbor information:

Router# show cdp neighbors

To display traffic statistics:

Router# sh cdp traffic

To display port and interface information:

Router# *sh cdp interface*

To disable CDP on an interface:

Router(config-if)# no cdp enable

To globally disable CDP:

Router(config)# no cdp run

Telnet

It is possible to telnet *from* a router into a remote device, using either the hostname or IP address of the remote device:

> Router# telnet Router2 **Router#** *telnet* 172.17.1.2

To exit a telnet session:

Router2# exit

To return to the router you telnetted *from*, without exiting the session:

- Hold CTRL+SHIFT+6 and then release
- Type the character x

To view all open telnet sessions:

Router# show sessions

Privilege Levels

IOS devices have a total of **16 privilege levels**, numbered 0 through 15. User Exec mode is privilege level 1. Privileged Exec mode is privilege level 15.

We can create a custom Privilege level, including the commands users are allowed to input at that mode:

> Router(config)# privilege exec all level 3 show interface **Router**(config)# privilege exec all level 3 show ip route Router(config)# privilege exec all level 3 show reload

To then enter that privilege level from User Mode:

Router> *enable 3*

Observing Performance Statistics on Cisco Routers

To view the processor load on a Cisco Router:

Router# show processes cpu

Part III

Routing

* * *

Section 13 - The Routing Table -

Routing Table Basics

Routing is the process of sending a packet of information from one *network* to another *network*. Thus, routes are usually based on the destination network, and not the destination host (host routes can exist, but are used only in rare circumstances).

To route, routers build Routing Tables that contain the following:

- The destination network and subnet mask
- The "next hop" router to get to the destination network
- Routing *metrics* and Administrative Distance

The routing table is concerned with two types of protocols:

- A **routed** protocol is a layer 3 protocol that applies logical addresses to devices and routes data between networks. Examples would be IP and IPX.
- A **routing** protocol dynamically builds the network, topology, and next hop information in routing tables. Examples would be RIP, IGRP, OSPF, etc.

To determine the *best* route to a destination, a router considers three elements (in this order):

- Prefix-Length
- **Metric** (*within* a routing protocol)
- Administrative Distance (between separate routing protocols)

Prefix-length is the number of bits used to identify the network, and is used to determine the most *specific* route. A longer prefix-length indicates a more specific route. For example, assume we are trying to reach a host address of 10.1.5.2/24. If we had routes to the following networks in the routing table:

> 10.1.5.0/24 10.0.0.0/8

The router will do a bit-by-bit comparison to find the most *specific* route (i.e., longest matching prefix). Since the 10.1.5.0/24 network is more specific, that route will be used, regardless of metric or Administrative Distance.

Administrative Distance vs. Metric

A "metric" allows a router to choose the best path within a routing protocol. Distance vector routing protocols use "distance" (usually hop-count) as their metric. Link state protocols utilize some sort of "cost" as their metric.

Only routes with the **best metric** are **added to the routing table**. Thus, even if a particular routing protocol (for example, RIP) has four routes to the same network, only the route with the best metric (hop-count in this example) would make it to the routing table. If multiple equal-metric routes exist to a particular network, most routing protocols will load-balance.

If your router is running multiple routing protocols, **Administrative Distance** is used to determine which routing protocol to *trust* the most. Lowest administrative distance wins.

Again: if a router receives two RIP routes to the same network, it will use the routes' **metric** to determine which path to use. If the metric is identical for both routes, the router will load balance between both paths.

If a router receives a RIP and an OSPF route to the same network, it will use **Administrative Distance** to determine which routing path to choose.

The Administrative Distance of common routing protocols (remember, lowest wins):

Connected	0
Static	1
EIGRP Summary	5
External BGP	20
Internal EIGRP	90
IGRP	100
OSPF	110
IS-IS	115
RIP	120
External EIGRP	170
Internal BGP	200
Unknown	255

A route with an "unknown" Administrative Distance will never be inserted into the routing table.

Viewing the routing table

The following command will allow you to view the routing table:

Router# *show ip route*

```
Gateway of last resort is 192.168.1.1 to network 0.0.0.0
     192.168.1.0/24 is directly connected, Ethernet0
     150.50.0.0/24 is subnetted, 1 subnets
     150.50.200.0 is directly connected, Loopback1
     192.168.123.0 is directly connected, Serial0
С
     192.168.111.0 is directly connected, Serial1
С
     10.0.0.0 [120/1] via 192.168.123.1, 00:00:00, Serial0
R
               [120/1] via 192.168.111.2, 00:00:00, Serial1
S*
     0.0.0.0/0 [1/0] via 192.168.1.1
```

Routes are labeled based on what protocol placed them in the table:

- C Directly connected
- S Static
- S* Default route
- o D EIGRP
- \circ R RIP
- \circ I IGRP
- \circ i IS-IS
- o O OSPF

Notice the RIP routes contain the following field: [120/1]. This indicates both the administrative distance and the metric (the 120 is the AD, and the 1 is the hop-count metric).

To clear all routes from the routing table, and thus forcing any routing protocol to repopulate the table:

Router# *clear ip route* *

Choosing the Best Route (Example)

Assume the following routes existed to the following host: 192.168.111.5/24

```
192.168.111.0/24 [110/58] via 192.168.131.1, 00:00:00, Serial3
     192.168.111.0/24 [120/1] via 192.168.123.1, 00:00:00, Serial0
     192.168.111.0/24 [120/5] via 192.168.5.2, 00:00:00, Serial1
R
     192.168.0.0/16 [1/0] via 10.1.1.1
```

We have two RIP routes, an OSPF route, and a Static route to that destination. Which route will be chosen by the router?

Remember the three criteria the router considers:

- Prefix-Length
- Metric
- Administrative Distance

The static route has the lowest administrative distance (1) of any of the routes; however, its **prefix-length** is less specific. 192.168.111.0/24 is a more specific route than 192.168.0.0/16. Remember, prefix-length is always considered first.

The second RIP route will not be inserted into the routing table, because it has a higher metric (5) than the first RIP route (1). Thus, our routing table will actually look as follows:

```
192.168.111.0/24 [110/58] via 192.168.131.1, 00:00:00, Serial3
0
     192.168.111.0/24 [120/1] via 192.168.123.1, 00:00:00, Serial0
R
     192.168.0.0/16 [1/0] via 10.1.1.1
```

Thus, the true choice is between the OSPF route and the first RIP route. OSPF has the lowest administrative distance, and thus that route will be preferred.

PLEASE NOTE: Calculating the lowest metric route within a routing protocol occurs before administrative distance chooses the route it "trusts" the most. This is why the order of the above "criteria" is prefix-length, metric, and *then* administrative distance.

However, the route with the lowest administrative distance is *always* preferred, regardless of metric (assuming the prefix-length is equal). Thus, the metric is *calculated* first, but not *preferred* first over AD.

Section 14 - Static vs. Dynamic Routing -

Static vs. Dynamic Routing

There are two basic methods of building a routing table: Statically or Dynamically.

A **static** routing table is created, maintained, and updated by a network administrator, manually. A static route to every network must be configured on every router for full connectivity. This provides a granular level of control over routing, but quickly becomes impractical on large networks.

Routers will *not* share static routes with each other, thus reducing CPU/RAM overhead and saving bandwidth. However, static routing is *not* fault-tolerant, as any change to the routing infrastructure (such as a link going down, or a new network added) requires manual intervention. Routers operating in a purely static environment cannot seamlessly choose a better route if a link becomes unavailable.

Static routes have an Administrative Distance (AD) of 1, and thus are always preferred over dynamic routes, unless the default AD is changed. A static route with an adjusted AD is called a **floating static route**, and is covered in greater detail in another guide.

A **dynamic** routing table is created, maintained, and updated by a *routing* protocol running on the router. Examples of routing protocols include RIP (Routing Information Protocol), **EIGRP** (Enhanced Interior Gateway Routing Protocol), and **OSPF** (Open Shortest Path First). Specific dynamic routing protocols are covered in great detail in other guides.

Routers do share dynamic routing information with each other, which increases CPU, RAM, and bandwidth usage. However, routing protocols are capable of dynamically choosing a different (or better) path when there is a change to the routing infrastructure.

Do not confuse *routing* protocols with *routed* protocols:

- A *routed* protocol is a Layer 3 protocol that applies logical addresses to devices and routes data between networks (such as IP)
- A *routing* protocol dynamically builds the network, topology, and next hop information in routing tables (such as RIP, EIGRP, etc.)

Static vs. Dynamic Routing (continued)

The following briefly outlines the advantages and disadvantages of *static* routing:

Advantages of Static Routing

- Minimal CPU/Memory overhead
- No bandwidth overhead (updates are not shared between routers)
- Granular control on how traffic is routed

Disadvantages of Static Routing

- Infrastructure changes must be manually adjusted
- No "dynamic" fault tolerance if a link goes down
- Impractical on large network

The following briefly outlines the advantages and disadvantages of *dynamic* routing:

Advantages of Dynamic Routing

- Simpler to configure on larger networks
- Will dynamically choose a different (or better) route if a link goes down
- Ability to load balance between multiple links
- Updates are shared between routers, thus consuming bandwidth

Disadvantages of Dynamic Routing

- Routing protocols put additional load on router CPU/RAM
- The choice of the "best route" is in the hands of the routing protocol, and not the network administrator

Dynamic Routing Categories

There are two distinct categories of dynamic routing protocols:

- Distance-vector protocols
- Link-state protocols

Examples of distance-vector protocols include **RIP** and **IGRP**. Examples of link-state protocols include **OSPF** and **IS-IS**.

EIGRP exhibits both distance-vector and link-state characteristics, and is considered a *hybrid* protocol.

Distance-vector Routing Protocols

All **distance-vector** routing protocols share several key characteristics:

- **Periodic** updates of the **full** routing table are sent to routing neighbors.
- Distance-vector protocols suffer from slow convergence, and are highly susceptible to loops.
- Some form of *distance* is used to calculate a route's metric.
- The **Bellman-Ford algorithm** is used to determine the shortest path.

A distance-vector routing protocol begins by advertising directly-connected networks to its neighbors. These updates are sent regularly (RIP – every 30 seconds; IGRP – every 90 seconds).

Neighbors will add the routes from these updates to their own routing tables. Each neighbor trusts this information *completely*, and will forward their full routing table (connected and learned routes) to every other neighbor. Thus, routers fully (and blindly) rely on neighbors for route information, a concept known as routing by rumor.

There are several disadvantages to this behavior. Because routing information is propagated from neighbor to neighbor via periodic updates, distance-vector protocols suffer from slow convergence. This, in addition to blind faith of neighbor updates, results in distance-vector protocols being highly susceptible to routing loops.

Distance-vector protocols utilize some form of **distance** to calculate a route's metric. RIP uses hopcount as its distance metric, and IGRP uses a composite of bandwidth and delay.

Link-State Routing Protocols

Link-state routing protocols were developed to alleviate the convergence and loop issues of distance-vector protocols. Link-state protocols maintain three separate tables:

- Neighbor table contains a list of all neighbors, and the interface each neighbor is connected off of. Neighbors are formed by sending Hello packets.
- **Topology table** otherwise known as the "link-state" table, contains a map of all links within an **area**, including each link's status.
- Shortest-Path table contains the *best* routes to each particular destination (otherwise known as the "routing" table")

Link-state protocols do *not* "route by rumor." Instead, routers send updates advertising the *state* of their *links* (a **link** is a directly-connected network). All routers know the state of all existing links within their **area**, and store this information in a **topology** table. All routers within an area have *identical* topology tables.

The best route to each link (network) is stored in the **routing** (or **shortest**path) table. If the state of a link changes, such as a router interface failing, an advertisement containing only this link-state change will be sent to all routers within that area. Each router will adjust its topology table accordingly, and will calculate a new best route if required.

By maintaining a consistent topology table among all routers within an area, link-state protocols can converge very quickly and are immune to routing loops.

Additionally, because updates are sent only during a link-state change, and contain *only* the change (and not the full table), link-state protocols are **less bandwidth intensive** than distance-vector protocols. However, the three link-state tables utilize more RAM and CPU on the router itself.

Link-state protocols utilize some form of **cost**, usually based on bandwidth, to calculate a route's metric. The **Dijkstra formula** is used to determine the shortest path.

Section 15 - Classful vs. Classless Routing -

Classful vs Classless routing protocols

Classful routing protocols do not send subnet mask information with their routing updates. A router running a classful routing protocol will react in one of two ways when receiving a route:

- If the router has a directly connected interface belonging to the same major network, it will apply the same subnet mask as that interface.
- If the router *does not* have any interfaces belonging to the same major network, it will apply the *classful* subnet mask to the route.

Belonging to same "major network" simply indicates that they belong to the same "classful" network. For example:

- 10.3.1.0 and 10.5.5.0 belong to the same major network (10.0.0.0)
- 10.1.4.5 and 11.1.4.4 do not belong to the same major network
- 192.168.1.1 and 192.168.1.254 belong to the same major network (192.168.1.0)
- 192.168.1.5 and 192.167.2.5 *do not* belong to the same major network.

Take the following example (assume the routing protocol is classful):



If Router B sends a routing update to Router A, it will *not* include the subnet mask for the 10.2.0.0 network. Thus, Router A must make a decision.

If Router A has a directly connected interface that belongs to the same major network (10.0.0.0), it will use the subnet mask of that interface for the route. For example, if Router A has an interface on the 10.4.0.0/16 network, it will apply a subnet mask of /16 to the 10.2.0.0 network.

If Router A *does not* have a directly connected interfacing belonging to the same major network, it will apply the classful subnet mask of /8. This can obviously cause routing difficulties.

When using classful routing protocols, the subnet mask must remain consistent throughout your entire network.

Classful vs Classless routing protocols (continued)

Classless routing protocols do send the subnet mask with their updates. Thus, Variable Length Subnet Masks (VLSMs) are allowed when using classless routing protocols.

Examples of *classful* routing protocols include RIPv1 and IGRP.

Examples of *classless* routing protocols include RIPv2, EIGRP, OSPF, and IS-IS.

The IP Classless Command

The preceding section described how classful and classless protocols differ when sending routing updates. Additionally, the router itself can operate either "classfully" or "classlessly" when actually routing data.

When a "classful" router has an interface connected to a major network, it believes it knows all routes connected to that major network.

For example, a router may have an interface attached to the 10.1.5.0/24 network. It may also have routes from a routing protocol, also for the 10.x.x.x network.

However, if the classful router receives a packet destined for a 10.x.x.x subnet that is *not* in the routing table, it will *drop* that packet, *even if there is* a default route.

Again, a classful router believes it knows all possible destinations in a major network.

To configure your router in "classful" mode:

Router(config)# *no ip classless*

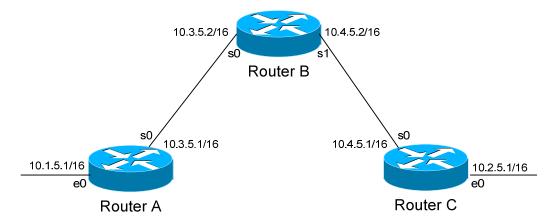
To configure your router in "classless" mode (this is default in IOS 12.0 and greater):

Router(config)# ip classless

(Reference: http://www.cisco.com/en/US/tech/tk365/technologies_tech_note09186a0080094823.shtml)

Limitations of Classful Routing Example

The following section will illustrate the limitations of classful routing, using RIPv1 as an example. Consider the following diagram:



This particular scenario will work when using RIPv1, despite the fact that we've subnetted the major 10.0.0.0 network. Notice that the subnets are contiguous (that is, they belong to the same major network), and use the same subnet mask.

When Router A sends a RIPv1 update to Router B via Serial0, it will not include the subnet mask for the 10.1.0.0 network. However, because the 10.3.0.0 network is in the same major network as the 10.1.0.0 network, it will **not summarize** the address. The route entry in the update will simply state "10.1.0.0".

Router B will accept this routing update, and realize that the interface receiving the update (Serial0) belongs to the same major network as the route entry of 10.1.0.0. It will then apply the subnet mask of its Serial0 interface to this route entry.

Router C will similarly send an entry for the 10.2.0.0 network to Router B. Router B's routing table will thus look like:

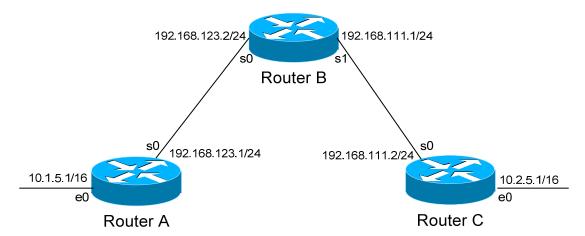
RouterB# *show ip route*

```
Gateway of last resort is not set
       10.0.0.0/16 is subnetted, 4 subnets
      10.3.0.0 is directly connected, Serial0
С
      10.4.0.0 is directly connected, Serial1
      10.1.0.0 [120/1] via 10.3.5.1, 00:00:00, Serial0
      10.2.0.0 [120/1] via 10.4.5.1, 00:00:00, Serial1
```

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Limitations of Classful Routing Example

Consider the following, slightly altered, example:



We'll assume that RIPv1 is configured correctly on all routers. Notice that our networks are no longer contiguous. Both Router A and Router C contain subnets of the 10.0.0.0 major network (10.1.0.0 and 10.2.0.0 respectively).

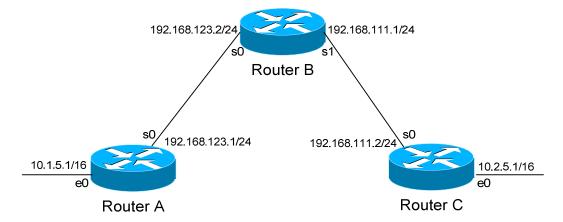
Separating these networks now are two Class C subnets (192.168.123.0 and 192.168.111.0).

Why is this a problem? Again, when Router A sends a RIPv1 update to Router B via Serial, it will not include the subnet mask for the 10.1.0.0 network. Instead, Router A will consider itself a **border** router, as the 10.1.0.0 and 192.168.123.0 networks *do not* belong to the same major network. Router A will **summarize** the 10.1.0.0/16 network to its classful boundary of 10.0.0.0/8.

Router B will accept this routing update, and realize that it does not have a directly connected interface in the 10.x.x.x scheme. Thus, it has no subnet mask to apply to this route. Because of this, Router B will install the summarized 10.0.0.0 route into its routing table.

Router C, similarly, will consider itself a border router between networks 10.2.0.0 and 192.168.111.0. Thus, Router C will also send a summarized 10.0.0.0 route to Router B.

Limitations of Classful Routing Example



Router B's routing table will then look like:

RouterB# show ip route

```
Gateway of last resort is not set
      192.168.123.0 is directly connected, Serial0
C.
      192.168.111.0 is directly connected, Serial1
      10.0.0.0 [120/1] via 192.168.123.1, 00:00:00, Serial0
               [120/1] via 192.168.111.2, 00:00:00, Serial1
```

That's right, Router B now has two equal metric routes to get to the summarized 10.0.0.0 network, one through Router A and the other through Router C. Router B will now *load balance* all traffic to *any* 10.x.x.x network between routers A and C. Suffice to say, this is not a good thing. ©

It gets better. Router B then tries to send routing updates to Router A and Router C, including the summary route of 10.0.0.0/8. Router A's routing table looks like:

RouterA# *show ip route*

```
Gateway of last resort is not set
      192.168.123.0 is directly connected, Serial0
С
      10.0.0.0/16 is subnetted, 1 subnet
      10.1.0.0 is directly connected, Ethernet0
```

Router A will receive the summarized 10.0.0.0/8 route from Router B, and will reject it. This is because it already has the summary network of 10.0.0.0 in its routing table, and it's directly connected. Router C will respond exactly the same, and the 10.1.0.0/16 and 10.2.0.0/16 networks will never be able to communicate.

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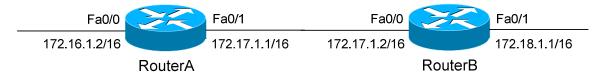
Section 16 - Configuring Static Routes -

Configuring Static Routes

The basic syntax for a static route is as follows:

Router(config)# ip route [destination_network] [subnet_mask] [next-hop]

Consider the following example:



Router A will have the 172.16.0.0/16 and 172.17.0.0/16 networks in its routing table as directly-connected routes. To add a static route on RouterA, pointing to the 172.18.0.0/16 network off of RouterB:

RouterA(config)# ip route 172.18.0.0 255.255.0.0 172.17.1.2

Notice that we point to the IP address on RouterB's fa0/0 interface as the next-hop address. Likewise, to add a static route on RouterB, pointing to the 172.16.0.0/16 network off of RouterA:

RouterB(config)# ip route 172.16.0.0 255.255.0.0 172.17.1.1

To remove a static route, simply type *no* in front of it:

RouterA(config)# no ip route 172.18.0.0 255.255.0.0 172.17.1.2

On point-to-point links, an exit-interface can be specified instead of a nexthop address. Still using the previous diagram as an example:

RouterA(config)# ip route 172.18.0.0 255.255.0.0 fa0/1

RouterB(config)# ip route 172.16.0.0 255.255.0.0 fa0/0

A static route using an exit-interface has an Administrative Distance of **0**, as opposed to the default AD of 1 for static routes. An exit-interface is only functional on a point-to-point link, as there is only one possible next-hop device.

Advanced Static Routes Parameters

The Administrative Distance of a static route can be changed to form a floating static route, which will only be used if there are no other routes with a lesser AD in the routing table. A floating static route is often used as a backup route to a dynamic routing protocol.

To change the Administrative Distance of a static route to 250:

RouterA(config)# ip route 172.18.0.0 255.255.0.0 172.17.1.2 250

Static routes will only remain in the routing table as long as the interface connecting to the next-hop router is up. To ensure a static route remains *permantly* in the routing table, even if the next-hop interface is down:

RouterA(config)# ip route 172.18.0.0 255.255.0.0 172.17.1.2 permanent

Static routes can additionally be used to *discard* traffic to specific networks, by directing that traffic to a virtual null interface:

RouterA(config)# ip route 10.0.0.0 255.0.0.0 null0

Default Routes

Normally, if a specific route to a particular network does not exist, a router will drop all traffic destined to that network.

A **default route**, or **gateway of last resort**, allows traffic to be forwarded, even without a specific route to a particular network.

The default route is identified by all zeros in both the network and subnet mask (0.0.0.0 0.0.0.0). It is the *least* specific route possible, and thus will only be used if a more specific route does not exist (hence "gateway of last resort").

To configure a default route:

RouterA(config)# ip route 0.0.0.0 0.0.0.0 172.17.1.2

Advanced default routing is covered in great detail in another guide.

Section 17 - RIP v1 & v2 -

RIP (Routing Information Protocol)

RIP is a standardized Distance Vector protocol, designed for use on smaller networks. RIP was one of the first true Distance Vector routing protocols, and is supported on a wide variety of systems.

RIP adheres to the following Distance Vector characteristics:

- RIP sends out periodic routing updates (every **30 seconds**)
- RIP sends out the full routing table every periodic update
- RIP uses a form of distance as its metric (in this case, **hopcount**)
- RIP uses the Bellman-Ford Distance Vector algorithm to determine the best "path" to a particular destination

Other characteristics of RIP include:

- RIP supports IP and IPX routing.
- RIP utilizes UDP port 520
- RIP routes have an administrative distance of 120.
- RIP has a maximum hopcount of 15 hops.

Any network that is 16 hops away or more is considered unreachable to RIP, thus the maximum diameter of the network is 15 hops. A metric of 16 hops in RIP is considered a poison route or infinity metric.

If multiple paths exist to a particular destination, RIP will load balance between those paths (by default, up to 4) only if the metric (hopcount) is equal. RIP uses a round-robin system of load-balancing between equal metric routes, which can lead to **pinhole congestion**.

For example, two paths might exist to a particular destination, one going through a 9600 baud link, the other via a T1. If the metric (hopcount) is equal, RIP will load-balance, sending an equal amount of traffic down the 9600 baud link and the T1. This will (obviously) cause the slower link to become congested.

<u>RIP Versions</u>

RIP has two versions, Version 1 (RIPv1) and Version 2 (RIPv2).

RIPv1 (RFC 1058) is **classful**, and thus does not include the subnet mask with its routing table updates. Because of this, RIPv1 does not support Variable Length Subnet Masks (VLSMs). When using RIPv1, networks must be contiguous, and subnets of a major network must be configured with identical subnet masks. Otherwise, route table inconsistencies (or worse) will occur.

RIPv1 sends updates as **broadcasts** to address 255.255.255.255.

RIPv2 (RFC 2543) is **classless**, and thus *does* include the subnet mask with its routing table updates. RIPv2 fully supports VLSMs, allowing discontiguous networks and varying subnet masks to exist.

Other enhancements offered by RIPv2 include:

- Routing updates are sent via **multicast**, using address **224.0.0.9**
- Encrypted authentication can be configured between RIPv2 routers
- Route tagging is supported (explained in a later section)

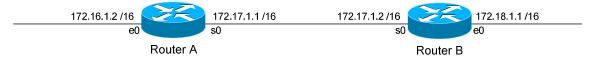
RIPv2 can interoperate with RIPv1. By default:

- RIPv1 routers will sent only Version 1 packets
- RIPv1 routers will receive both Version 1 and 2 updates
- RIPv2 routers will both send and receive only Version 2 updates

We can control the version of RIP a particular interface will "send" or "receive."

Unless RIPv2 is manually specified, a Cisco will default to RIPv1 when configuring RIP.

RIPv1 Basic Configuration



Routing protocol configuration occurs in Global Configuration mode. On Router A, to configure RIP, we would type:

> **Router**(config)# router rip Router(config-router)# network 172.16.0.0 Router(config-router)# network 172.17.0.0

The first command, *router rip*, enables the RIP process.

The *network* statements tell RIP which networks you wish to advertise to other RIP routers. We simply list the networks that are directly connected to our router. Notice that we specify the networks at their classful boundaries, and we do not specify a subnet mask.

To configure Router B:

Router(config)# router rip Router(config-router)# network 172.17.0.0 Router(config-router)# network 172.18.0.0

The routing table on Router A will look like:

RouterA# show ip route

```
<eliminated irrelevant header>
Gateway of last resort is not set
      172.16.0.0 is directly connected, Ethernet0
С
      172.17.0.0 is directly connected, Serial0
      172.18.0.0 [120/1] via 172.17.1.2, 00:00:00, Serial0
```

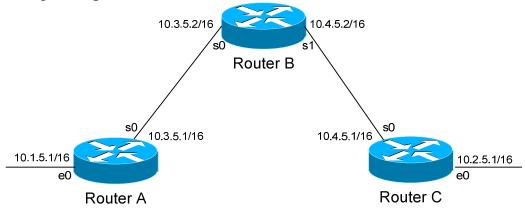
The routing table on Router B will look like:

RouterB# show ip route

```
<eliminated irrelevant header>
Gateway of last resort is not set
      172.17.0.0 is directly connected, Serial0
      172.18.0.0 is directly connected, Ethernet0
      172.16.0.0 [120/1] via 172.17.1.1, 00:00:00, Serial0
```

Limitations of RIPv1

The example on the previous page works fine with RIPv1, because the networks are contiguous and the subnet masks are consistent. Consider the following example:



This particular scenario will *still* work when using RIPv1, despite the fact that we've subnetted the major 10.0.0.0 network. Notice that the subnets are contiguous (that is, they belong to the same major network), and use the same subnet mask.

When Router A sends a RIPv1 update to Router B via Serial0, it will not include the subnet mask for the 10.1.0.0 network. However, because the 10.3.0.0 network is in the same major network as the 10.1.0.0 network, it will **not summarize** the address. The route entry in the update will simply state "10.1.0.0".

Router B will accept this routing update, and realize that the interface receiving the update (Serial0) belongs to the same major network as the route entry of 10.1.0.0. It will then apply the subnet mask of its Serial0 interface to this route entry.

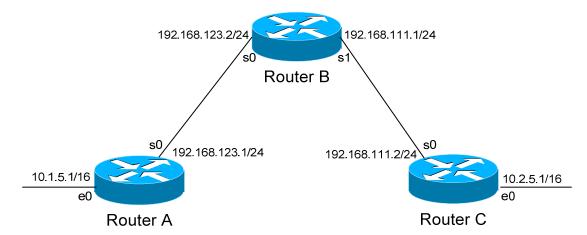
Router C will similarly send an entry for the 10.2.0.0 network to Router B. Router B's routing table will thus look like:

RouterB# *show ip route*

```
Gateway of last resort is not set
       10.0.0.0/16 is subnetted, 4 subnets
      10.3.0.0 is directly connected, Serial0
С
      10.4.0.0 is directly connected, Serial1
      10.1.0.0 [120/1] via 10.3.5.1, 00:00:00, Serial0
R
      10.2.0.0 [120/1] via 10.4.5.1, 00:00:00, Serial1
```

Limitations of RIPv1 (continued)

Consider the following, slightly altered, example:



We'll assume that RIPv1 is configured correctly on all routers. Notice that our networks are no longer contiguous. Both Router A and Router C contain subnets of the 10.0.0.0 major network (10.1.0.0 and 10.2.0.0 respectively).

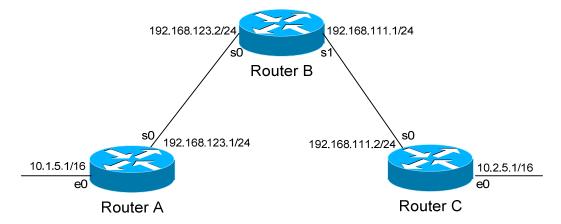
Separating these networks now are two Class C subnets (192.168.123.0 and 192.168.111.0).

Why is this a problem? Again, when Router A sends a RIPv1 update to Router B via Serial, it will not include the subnet mask for the 10.1.0.0 network. Instead, Router A will consider itself a **border** router, as the 10.1.0.0 and 192.168.123.0 networks *do not* belong to the same major network. Router A will **summarize** the 10.1.0.0/16 network to its classful boundary of 10.0.0.0/8.

Router B will accept this routing update, and realize that it does not have a directly connected interface in the 10.x.x.x scheme. Thus, it has no subnet mask to apply to this route. Because of this, Router B will install the summarized 10.0.0.0 route into its routing table.

Router C, similarly, will consider itself a border router between networks 10.2.0.0 and 192.168.111.0. Thus, Router C will also send a summarized 10.0.0.0 route to Router B.

Limitations of RIPv1 (continued)



Router B's routing table will then look like:

RouterB# show ip route

```
Gateway of last resort is not set
      192.168.123.0 is directly connected, Serial0
C.
      192.168.111.0 is directly connected, Serial1
      10.0.0.0 [120/1] via 192.168.123.1, 00:00:00, Serial0
               [120/1] via 192.168.111.2, 00:00:00, Serial1
```

That's right, Router B now has two equal metric routes to get to the summarized 10.0.0.0 network, one through Router A and the other through Router C. Router B will now *load balance* all traffic to *any* 10.x.x.x network between routers A and C. Suffice to say, this is not a good thing. ©

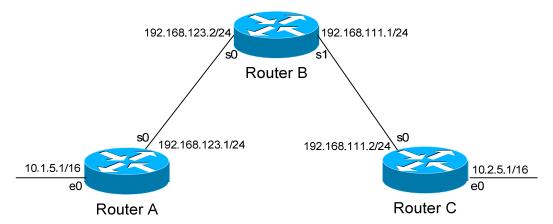
It gets better. Router B then tries to send routing updates to Router A and Router C, including the summary route of 10.0.0.0/8. Router A's routing table looks like:

RouterA# *show ip route*

```
Gateway of last resort is not set
      192.168.123.0 is directly connected, Serial0
С
      10.0.0.0/16 is subnetted, 1 subnet
      10.1.0.0 is directly connected, Ethernet0
```

Router A will receive the summarized 10.0.0.0/8 route from Router B, and will reject it. This is because it already has the summary network of 10.0.0.0 in its routing table, and it's directly connected. Router C will respond exactly the same, and the 10.1.0.0/16 and 10.2.0.0/16 networks will never be able to communicate.

RIPv2 Configuration



RIPv2 overcomes the limitations of RIPv1 by including the subnet mask in its routing updates. By default, Cisco routers will use RIPv1. To change to Version 2, you must type:

> **Router**(config)# router rip **Router(config-router)**# version 2

Thus, the configuration of Router A would be:

RouterA(config)# router rip RouterA(config-router)# version 2 RouterA(config-router)# network 10.0.0.0 RouterA(config-router)# network 192.168.123.0

Despite the fact that RIPv2 is a classless routing protocol, we still specify networks at their classful boundaries, without a subnet mask.

However, when Router A sends a **RIPv2** update to Router B via Serial0, by default it will still **summarize** the 10.1.0.0/16 network to 10.0.0.0/8. Again, this is because the 10.1.0.0 and 192.168.123.0 networks do not belong to the same major network. Thus, RIPv2 acts like RIPv1 in this circumstance...

...unless you disable auto summarization:

RouterA(config)# router rip RouterA(config-router)# version 2 **RouterA**(config-router)# no auto-summary

The *no auto-summary* command will prevent Router A from summarizing the 10.1.0.0 network. Instead, Router A will send an update that includes both the subnetted network (10.1.0.0) and its subnet mask (255.255.0.0).

<u>RIP Timers</u>

RIP has four basic timers:

Update Timer (default **30 seconds**) – indicates how often the router will send out a routing table update.

Invalid Timer (default **180 seconds**) – indicates how long a route will remain in a routing table before being marked as invalid, if no new updates are heard about this route. The invalid timer will be reset if an update is received for that particular route *before* the timer expires.

A route marked as invalid is *not* immediately removed from the routing table. Instead, the route is marked (and advertised) with a metric of 16, indicating it is unreachable, and placed in a **hold-down** state.

Hold-down Timer (default **180 seconds**) – indicates how long RIP will "suppress" a route that it has placed in a **hold-down** state. RIP will not accept any new updates for routes in a hold-down state, until the hold-down timer expires.

A route will enter a hold-down state for one of three reasons:

- The invalid timer has expired.
- An update has been received from another router, marking that route with a metric of 16 (unreachable).
- An update has been received from another router, marking that route with a higher metric than what is currently in the routing table (this is to prevent loops).

Flush Timer (default 240 seconds) – indicates how long a route can remain in a routing table before being flushed, if no new updates are heard about this route. The flush timer runs **concurrently** with the hold-down timer, and thus will flush out a route 60 seconds after it has been marked invalid.

RIP timers must be identical on all routers on the RIP network, otherwise massive instability will occur.

RIP Timers Configuration and Example



Consider the above example. Router A receives a RIP update from Router B that includes network 172.18.0.0. Router A adds this network to its routing table:

RouterA# show ip route

```
Gateway of last resort is not set
      172.16.0.0 is directly connected, Ethernet0
      172.17.0.0 is directly connected, Serial0
      172.18.0.0 [120/1] via 172.17.1.2, 00:00:00, Serial0
```

Immediately, Router A sets an **invalid** timer of 180 seconds. If no update for this route is heard for 180 seconds, several things will occur:

- The route is marked as invalid in the routing table.
- The route enters a **hold-down** state (triggering the hold-down timer).
- The flush timer is triggered.
- The route is advertised to all other routers as unreachable.

The hold-down timer runs for 180 seconds after the route is marked as invalid. The router will not accept any new updates for this route until this hold-down period expires. The **flush** timer runs for 240 seconds after the route is marked as invalid.

If no update is heard at all, the route will be deleted completely once the flush timer expires. By default, this will be 60 seconds after the hold-down timer expires (240 - 180 seconds) = 60 seconds. Remember, the hold-down and flush timers run concurrently.

To configure the RIP timers:

Router(config)# router rip Router(config-router)# timers basic 20 120 120 160

The timers basic command allows us to change the update (20), invalid (120), hold-down (120), and flush (240) timers. To return the timers back to their defaults:

Router(config-router)# *no timers basic*

RIP Loop Avoidance Mechanisms

RIP, as a Distance Vector routing protocol, is susceptible to loops.



Let's assume no loop avoidance mechanisms are configured on either router. If the 172.18.0.0 network fails, Router B will send out an update to Router A within 30 seconds (whenever its update timer expires) stating that route is unreachable (metric = 16).

But what if an update from Router A reaches Router B before this can happen? Router A believes it can reach the 172.18.0.0 network in one hop (through Router B). This will cause Router B to believe it can reach the failed 172.18.0.0 network in **two hops**, through Router A. Both routers will continue to increment the metric for the network until they reach a hop count of **16**, which is unreachable. This behavior is known as **counting to infinity**.

How can we prevent this from happening? There are several loop avoidance mechanisms:

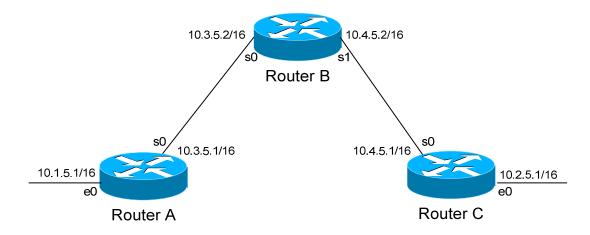
Split-Horizon – Prevents a routing update from being sent out the interface it was received on. In our above example, this would prevent Router A from sending an update for the 172.18.0.0 network back to Router B, as it originally learned the route from Router B. Split-horizon is enabled by default on Cisco Routers.

Route-Poisoning – Works in conjunction with split-horizon, by **triggering** an automatic update for the failed network, without waiting for the update timer to expire. This update is sent out all interfaces with an infinity metric for that network.

Hold-Down Timers – Prevents RIP from accepting any new updates for routes in a hold-down state, until the hold-down timer expires. If Router A sends an update to Router B with a *higher* metric than what is currently in Router B's routing table, that route will be placed in a hold-down state. (Router A's metric for the 172.18.0.0 network is 1; while Router B's metric is 0).

RIP Passive Interfaces

It is possible to control which router interfaces will participate in the RIP process.



Consider the following scenario. Router C does not want to participate in the RIP domain. However, it still wants to *listen* to updates being sent from Router B, just not *send* any updates back to Router B:

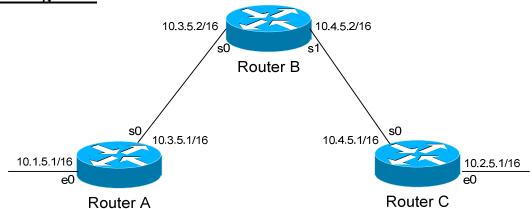
> RouterC(config)# router rip RouterC(config-router)# network 10.4.0.0 RouterC(config-router)# network 10.2.0.0 **RouterC(config-router)**# passive-interface s0

The passive-interface command will prevent updates from being sent out of the SerialO interface, but Router C will still receive updates on this interface.

We can configure **all** interfaces to be passive using the *passive-interface* default command, and then individually use the no passive-interface command on the interfaces we **do** want updates to be sent out:

> RouterC(config)# router rip RouterC(config-router)# network 10.4.0.0 RouterC(config-router)# network 10.2.0.0 **RouterC(config-router)**# passive-interface default **RouterC(config-router)**# no passive-interface e0

RIP Neighbors



Recall that RIPv1 sends out its updates as broadcasts, whereas RIPv2 sends out its updates as **multicasts** to the 224.0.0.9 address. We can configure specific RIP *neighbor* commands, which will allow us to **unicast** routing updates to those neighbors.

On Router B:

RouterB(config)# router rip RouterB(config-router)# network 10.3.0.0 RouterB(config-router)# network 10.4.0.0 RouterB(config-router)# neighbor 10.3.5.1 **RouterB**(config-router)# neighbor 10.4.5.1

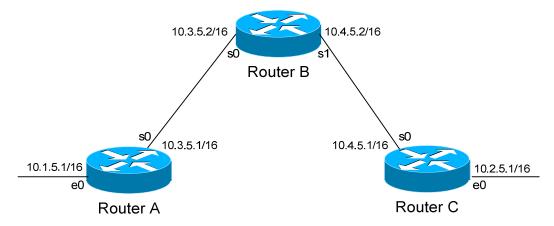
Router B will now unicast RIP updates to Router A and Router C.

However, Router B will still broadcast (if RIPv1) or multicast (if RIPv2) its updates, in addition to sending unicast updates to its neighbors. In order to prevent broadcast/multicast updates, we must also use **passive interfaces**:

> RouterB(config)# router rip RouterB(config-router)# passive-interface s0 **RouterB(config-router)**# passive-interface s1 RouterB(config-router)# neighbor 10.3.5.1 **RouterB**(config-router)# neighbor 10.4.5.1

The passive-interface commands prevent the updates from being broadcasted or multicasted. The *neighbor* commands still allow unicast updates to those specific neighbors.

Interoperating between RIPv1 and RIPv2



Recall that, with some configuration, RIPv1 and RIPv2 can interoperate. By default:

- RIPv1 routers will sent only Version 1 packets
- RIPv1 routers will receive both Version 1 and 2 updates
- RIPv2 routers will both send and receive only Version 2 updates

If Router A is running RIP v1, and Router B is running RIP v2, some additional configuration is necessary.

Either we must configure Router A to send Version 2 updates:

RouterA(config)# interface s0 **RouterA**(config-if)# ip rip send version 2

Or configure Router B to accept Version 1 updates.

RouterB(config)# interface s0 **RouterB**(config-if)# ip rip receive version 1

Notice that this is configured on an interface. Essentially, we're configuring the version of RIP on a per-interface basis.

We can also have an interface send or receive both versions simultaneously:

RouterB(config)# interface s0 **RouterB**(config-if)# ip rip receive version 1 2

We can further for RIPv2 to send broadcast updates, instead of multicasts:

RouterB(config)# interface s0 **RouterB(config)**# *ip rip v2-broadcast*

Triggering RIP Updates

On point-to-point interfaces, we can actually force RIP to only send routing updates if there is a change:

```
RouterB(config)# interface s0.150 point-to-point
RouterB(config-if)# ip rip triggered
```

Again, this is only applicable to **point-to-point** links. We cannot configure RIP triggered updates on an Ethernet network.

Troubleshooting RIP

Various troubleshooting commands exist for RIP.

To view the IP routing table:

Router# *show ip route*

```
<eliminated irrelevant header>
Gateway of last resort is not set
      172.16.0.0 is directly connected, Ethernet0
      172.17.0.0 is directly connected, Serial0
     172.18.0.0 [120/1] via 172.17.1.2, 00:00:15, Serial0
     192.168.123.0 [120/1] via 172.16.1.1, 00:00:00, Ethernet0
```

To view a specific route within the IP routing table:

Router# show ip route 172.18.0.0

```
Routing entry for 172.18.0.0/16
      Known via "rip", distance 120, metric 1
      Last update from 172.17.1.2 on Serial 0, 00:00:15 ago
```

To debug RIP in real time:

Router# debug ip rip

Troubleshooting RIP (continued)

To view information specific to the RIP protocol:

Router# *show ip protocols*

```
Routing Protocol is "rip"
  Sending updates every 30 seconds, next due in 20 seconds
  Invalid after 180 seconds, hold down 180, flushed after 240
  Outgoing update filter list for all interfaces is not set
  Incoming update filter list for all interfaces is not set
  Incoming routes will have 4 added to metric if on list 1
  Redistributing: connected, static, rip
  Default version control: send version 1, receive any version
   Interface
                         Send Recv Triggered RIP Key-chain
   Ethernet0
Serial0
                         1 1 2
                         1 2 1 2
  Automatic network summarization is in effect
  Maximum path: 4
  Routing for Networks:
    172.16.0.0
   172.17.0.0
  Routing Information Sources:
   Gateway Distance Last Update 172.17.1.2 120 00:00:17
  Distance: (default is 120)
```

This command provides us with information on RIP timers, on the RIP versions configured on each interface, and the specific networks RIP is advertising.

To view all routes in the RIP database, and not just the entries added to the routing table:

Router# *show ip rip database*

```
7.0.0.0/8 auto-summary
7.0.0.0/8
     [5] via 172.16.1.1, 00:00:06, Ethernet0
172.16.0.0/16 directly connected, Ethernet0 172.17.0.0/16 directly connected, Serial0
```

Section 18 - Interior Gateway Routing Protocol -

IGRP (Interior Gateway Routing Protocol)

IGRP is a Cisco-proprietary Distance-Vector protocol, designed to be more scalable than RIP, its standardized counterpart.

IGRP adheres to the following Distance-Vector characteristics:

- IGRP sends out periodic routing updates (every **90 seconds**).
- IGRP sends out the full routing table every periodic update.
- IGRP uses a form of distance as its metric (in this case, a composite of **bandwidth** and **delay**).
- IGRP uses the Bellman-Ford Distance Vector algorithm to determine the best "path" to a particular destination.

Other characteristics of IGRP include:

- IGRP supports **only IP routing**.
- IGRP utilizes **IP protocol 9**.
- IGRP routes have an administrative distance of 100.
- IGRP, by default, supports a maximum of **100 hops.** This value can be adjusted to a maximum of **255 hops**.
- IGRP is a **classful** routing protocol.

IGRP uses **Bandwidth** and **Delay of the Line**, by default, to calculate its distance metric. Reliability, Load, and MTU are optional attributes that can be used to calculate the distance metric.

IGRP requires that you include an Autonomous System (AS) number in its configuration. Only routers in the same Autonomous system will send updates between each other.

Configuring IGRP



Routing protocol configuration occurs in Global Configuration mode. On Router A, to configure IGRP, we would type:

> **Router(config)#** router igrp 10 **Router(config-router)**# network 172.16.0.0 Router(config-router)# network 172.17.0.0

The first command, router igrp 10, enables the IGRP process. The "10" indicates the Autonomous System number that we are using. Only other IGRP routers in Autonomous System 10 will share updates with this router.

The *network* statements tell IGRP which networks you wish to advertise to other RIP routers. We simply list the networks that are directly connected to our router. Notice that we specify the networks at their classful boundaries, and we do not specify a subnet mask.

To configure Router B:

Router(config)# router igrp 10 Router(config-router)# network 172.17.0.0 Router(config-router)# network 172.18.0.0

The routing table on Router A will look like:

RouterA# *show ip route*

```
Gateway of last resort is not set
С
      172.16.0.0 is directly connected, Ethernet0
      172.17.0.0 is directly connected, Serial0
      172.18.0.0 [120/1] via 172.17.1.2, 00:00:00, Serial0
```

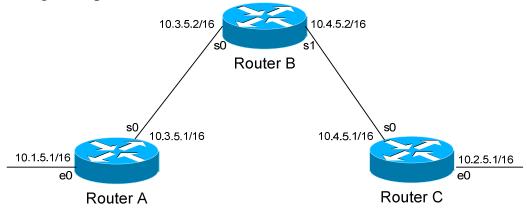
The routing table on Router B will look like:

RouterB# show ip route

```
Gateway of last resort is not set
      172.17.0.0 is directly connected, Serial0
      172.18.0.0 is directly connected, Ethernet0
      172.16.0.0 [120/1] via 172.17.1.1, 00:00:00, Serial0
```

Limitations of IGRP

The example on the previous page works fine with IGRP, because the networks are contiguous and the subnet masks are consistent. Consider the following example:



This particular scenario will *still* work when using IGRP, despite the fact that we've subnetted the major 10.0.0.0 network. Notice that the subnets are contiguous (that is, they belong to the same major network), and use the same subnet mask.

When Router A sends an IGRP update to Router B via Serial0, it will not include the subnet mask for the 10.1.0.0 network. However, because the 10.3.0.0 network is in the same major network as the 10.1.0.0 network, it will **not summarize** the address. The route entry in the update will simply state "10.1.0.0".

Router B will accept this routing update, and realize that the interface receiving the update (Serial0) belongs to the same major network as the route entry of 10.1.0.0. It will then apply the subnet mask of its Serial0 interface to this route entry.

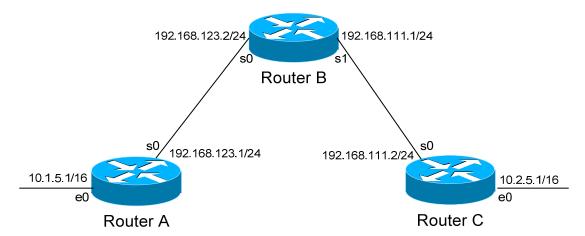
Router C will similarly send an entry for the 10.2.0.0 network to Router B. Router B's routing table will thus look like:

RouterB# *show ip route*

```
Gateway of last resort is not set
       10.0.0.0/16 is subnetted, 4 subnets
      10.3.0.0 is directly connected, Serial0
С
      10.4.0.0 is directly connected, Serial1
      10.1.0.0 [120/1] via 10.3.5.1, 00:00:00, Serial0
Ι
      10.2.0.0 [120/1] via 10.4.5.1, 00:00:00, Serial1
```

Limitations of IGRP (continued)

Consider the following, slightly altered, example:



We'll assume that IGRP is configured correctly on all routers. Notice that our networks are no longer contiguous. Both Router A and Router C contain subnets of the 10.0.0.0 major network (10.1.0.0 and 10.2.0.0 respectively).

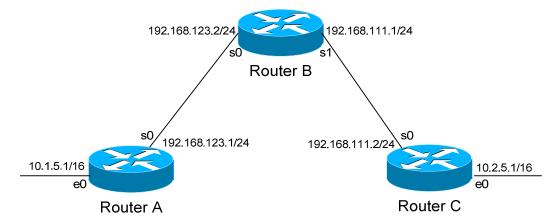
Separating these networks now are two Class C subnets (192.168.123.0 and 192.168.111.0).

Why is this a problem? Again, when Router A sends an IGRP update to Router B via Serial, it will not include the subnet mask for the 10.1.0.0 network. Instead, Router A will consider itself a **border** router, as the 10.1.0.0 and 192.168.123.0 networks *do not* belong to the same major network. Router A will **summarize** the 10.1.0.0/16 network to its classful boundary of 10.0.0.0/8.

Router B will accept this routing update, and realize that it does not have a directly connected interface in the 10.x.x.x scheme. Thus, it has no subnet mask to apply to this route. Because of this, Router B will install the summarized 10.0.0.0 route into its routing table.

Router C, similarly, will consider itself a border router between networks 10.2.0.0 and 192.168.111.0. Thus, Router C will also send a summarized 10.0.0.0 route to Router B.

Limitations of IGRP (continued)



Router B's routing table will then look like:

RouterB# show ip route

```
Gateway of last resort is not set
      192.168.123.0 is directly connected, Serial0
      192.168.111.0 is directly connected, Serial1
      10.0.0.0 [120/1] via 192.168.123.1, 00:00:00, Serial0
               [120/1] via 192.168.111.2, 00:00:00, Serial1
```

That's right, Router B now has two equal metric routes to get to the summarized 10.0.0.0 network, one through Router A and the other through Router C. Router B will now *load balance* all traffic to *any* 10.x.x.x network between routers A and C. Suffice to say, this is not a good thing. ©

It gets better. Router B then tries to send routing updates to Router A and Router C, including the summary route of 10.0.0.0/8. Router A's routing table looks like:

RouterA# *show ip route*

```
Gateway of last resort is not set
      192.168.123.0 is directly connected, Serial0
С
      10.0.0.0/16 is subnetted, 1 subnet
      10.1.0.0 is directly connected, Ethernet0
```

Router A will receive the summarized 10.0.0.0/8 route from Router B, and will reject it. This is because it already has the summary network of 10.0.0.0 in its routing table, and it's directly connected. Router C will respond exactly the same, and the 10.1.0.0/16 and 10.2.0.0/16 networks will never be able to communicate.

<u>IGRP Timers</u>

IGRP has four basic timers:

Update Timer (default **90 seconds**) – indicates how often the router will send out a routing table update.

Invalid Timer (default **270 seconds**) – indicates how long a route will remain in a routing table before being marked as invalid, if no new updates are heard about this route. The invalid timer will be reset if an update is received for that particular route *before* the timer expires.

A route marked as invalid is *not* immediately removed from the routing table. Instead, the route is marked (and advertised) with a metric of 101 (remember, 100 maximum hops is default), indicating it is unreachable, and placed in a **hold-down** state.

Hold-down Timer (default **280 seconds**) – indicates how long IGRP will "suppress" a route that it has placed in a **hold-down** state. IGRP will not accept any new updates for routes in a hold-down state, until the hold-down timer expires.

A route will enter a hold-down state for one of three reasons:

- The invalid timer has expired.
- An update has been received from another router, marking that route with a metric of 101 (unreachable).
- An update has been received from another router, marking that route with a *higher* metric than what is currently in the routing table (this is to prevent loops).

Flush Timer (default 630 seconds) – indicates how long a route can remain in a routing table before being flushed, if no new updates are heard about this route. The flush timer runs **concurrently** with the hold-down timer, and thus will flush out a route 350 seconds after it has been marked invalid.

IGRP timers must be identical on **all** routers on the IGRP network, otherwise massive instability will occur.

IGRP Loop Avoidance Mechanisms

IGRP, as a Distance Vector routing protocol, is susceptible to loops.



Let's assume no loop avoidance mechanisms are configured on either router. If the 172.18.0.0 network fails, Router B will send out an update to Router A within 30 seconds (whenever its update timer expires) stating that route is unreachable.

But what if an update from Router A reaches Router B before this can happen? Router A believes it can reach the 172.18.0.0 network in one hop (through Router B). This will cause Router B to believe it can reach the failed 172.18.0.0 network in **two hops**, through Router A. Both routers will continue to increment the metric for the network until they reach an infinity hop count (by default, 101). This behavior is known as **counting to infinity**.

How can we prevent this from happening? There are several loop avoidance mechanisms:

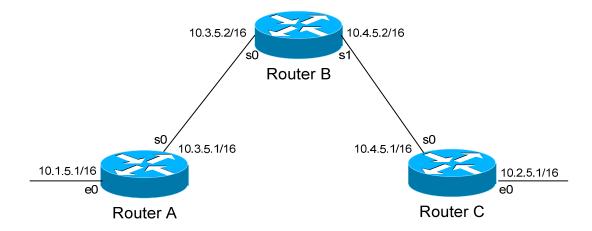
Split-Horizon – Prevents a routing update from being sent out the interface it was received on. In our above example, this would prevent Router A from sending an update for the 172.18.0.0 network back to Router B, as it originally learned the route from Router B. Split-horizon is enabled by default on Cisco Routers.

Route-Poisoning – Works in conjunction with split-horizon, by **triggering** an automatic update for the failed network, without waiting for the update timer to expire. This update is sent out all interfaces with an infinity metric for that network.

Hold-Down Timers – Prevents IGRP from accepting any new updates for routes in a hold-down state, until the hold-down timer expires. If Router A sends an update to Router B with a higher metric than what is currently in Router B's routing table, that route will be placed in a hold-down state.

IGRP Passive Interfaces

It is possible to control which router interfaces will participate in the IGRP process.



Consider the following scenario. Router C does not want to participate in the IGRP domain. However, it still wants to *listen* to updates being sent from Router B, just not *send* any updates back to Router B:

> RouterC(config)# router igrp 10 RouterC(config-router)# network 10.4.0.0 RouterC(config-router)# network 10.2.0.0 **RouterC**(config-router)# passive-interface s0

The passive-interface command will prevent updates from being sent out of the SerialO interface, but Router C will still receive updates on this interface.

We can configure **all** interfaces to be passive using the *passive-interface* default command, and then individually use the no passive-interface command on the interfaces we **do** want updates to be sent out:

> **RouterC(config)#** router igrp 10 RouterC(config-router)# network 10.4.0.0 RouterC(config-router)# network 10.2.0.0 **RouterC(config-router)**# passive-interface default **RouterC(config-router)**# no passive-interface e0

Advanced IGRP Configuration

To change the maximum hop-count to 255 for IGRP:

Router(config)# router igrp 10 Router(config-router)# metric maximum-hops 255

Section 19

- Enhanced Interior Gateway Routing Protocol -

EIGRP (Enhanced Interior Gateway Routing Protocol)

EIGRP is a Cisco-proprietary Hybrid routing protocol, incorporating features of both Distance-Vector and Link-State routing protocols.

EIGRP adheres to the following Hybrid characteristics:

- EIGRP uses **Diffusing Update Algorithm (DUAL)** to determine the best path among all "feasible" paths. DUAL also helps ensure a loopfree routing environment.
- EIGRP will form **neighbor** relationships with adjacent routers in the same Autonomous System (AS).
- EIGRP traffic is either sent as unicasts, or as multicasts on address **224.0.0.10**, depending on the EIGRP packet type.
- Reliable Transport Protocol (RTP) is used to ensure delivery of most EIGRP packets.
- EIGRP routers **do not** send periodic, full-table routing updates. Updates are sent when a change occurs, and include *only* the change.
- EIGRP is a classless protocol, and thus supports VLSMs.

Other characteristics of EIGRP include:

- EIGRP supports IP, IPX, and Appletalk routing.
- EIGRP applies an Administrative Distance of 90 for routes originating within the local Autonomous System.
- EIGRP applies an Administrative Distance of 170 for external routes coming from *outside* the local Autonomous System
- EIGRP uses **Bandwidth** and **Delay of the Line**, by default, to calculate its distance metric. It also supports three other parameters to calculate its metric: Reliability, Load, and MTU.
- EIGRP has a maximum hop-count of 224, though the default maximum hop-count is set to 100.

EIGRP, much like OSPF, builds three separate tables:

- Neighbor table list of all neighboring routers. Neighbors must belong to the same Autonomous System
- **Topology table** list of *all* routes in the Autonomous System
- **Routing table** contains the *best* route for each known network

EIGRP Neighbors

EIGRP forms neighbor relationships, called **adjacencies**, with other routers in the same AS by exchanging **Hello** packets. Only after an adjacency is formed can routers share routing information. Hello packets are sent as multicasts to address 224.0.0.10.

By default, on LAN and high-speed WAN interfaces, EIGRP Hellos are sent every 5 seconds. On slower WAN links (T1 speed or slower), EIGRP Hellos are sent every **60 seconds** by default.

The EIGRP Hello timer can be adjusted on a per interface basis:

Router(config-if)# ip hello-interval eigrp 10 7

The above command allows us to change the *hello* timer to 7 seconds for Autonomous System 10.

In addition to the Hello timer, EIGRP neighbors are stamped with a **Hold** timer. The Hold timer indicates how long a router should wait before marking a neighbor inactive, if it stops receiving hello packets from that neighbor.

By default, the Hold timer is **three times** the Hello timer. Thus, on highspeed links the timer is set to 15 seconds, and on slower WAN links the timer is set to **180 seconds**.

The Hold timer can also be adjusted on a *per interface* basis:

Router(config-if)# ip hold-interval eigrp 10 21

The above command allows us to change the *hold* timer to 21 seconds for Autonomous System 10.

Changing the Hello timer **does not** automatically change the Hold timer. Additionally, Hello and Hold timers *do not* need to match between routers for an EIGRP neighbor relationship to form.

EIGRP Neighbors (continued)

A **neighbor table** is constructed from the EIGRP Hello packets, which includes the following information:

- The IP address of the neighboring router.
- The local interface that received the neighbor's Hello packet.
- The Hold timer.
- A sequence number indicating the order neighbors were learned.

Adjacencies will not form unless the **primary IP addresses** on connecting interfaces are on the same subnet. Neighbors *cannot* be formed on secondary addresses.

If connecting interfaces are on *different* subnets, an EIGRP router will log the following error to console when a multicast Hello is received:

```
00:11:22: IP-EIGRP: Neighbor 172.16.1.1 not on common
subnet for Serial0
```

Always ensure that primary IP addresses belong to the same subnet between EIGRP neighbors.

To log all neighbor messages and errors to console, use the following two commands:

```
Router(config)# router eigrp 10
Router(config-router)# eigrp log-neighbor-changes
Router(config-router)# eigrp log-neighbor-warnings
```

(Reference: http://www.cisco.com/en/US/tech/tk365/technologies_configuration_example09186a0080093f09.shtml)

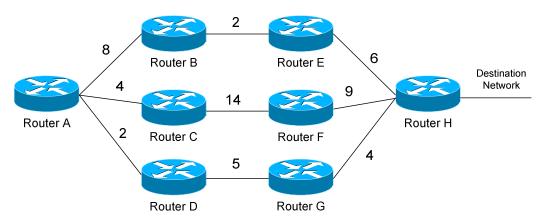
The EIGRP Topology Table

Once EIGRP neighbors form adjacencies, they will begin to share routing information. Each router's update contains a list of all routes known by that router, and the respective metrics for those routes.

All such routes are added to an EIGRP router's topology table. The route with the lowest metric to each network will become the **Feasible Distance** (FD). The Feasible Distance for each network will be installed into the routing table.

The Feasible Distance is derived from the **Advertised Distance** of the router sending the update, and the local router's metric to the advertising router.

Confused? Consider the following example:



Router A has three separate paths to the Destination Network, either through Router B, C, or D. If we add up the metrics to form a "distance" (the metrics are greatly simplified in this example), we can determine the following:

- Router B's Feasible Distance to the Destination Network is 8.
- Router C's Feasible Distance to the Destination Network is 23.
- Router D's Feasible Distance to the Destination Network is 9.

Router B sends an update to Router A, it will provide an Advertised **Distance** of 8 to the Destination Network. Router C will provide an AD of 23, and D will provide an AD of 9.

Router A calculates the total distance to the Destination network by adding the AD of the advertising router, with its own distance to reach that advertising router. For example, Router A's metric to Router B is 8; thus, the total distance will be 16 to reach the Destination Network through Router B.

Router B Router E Destination Network 14 Router A Router H 2 Router C Router F 4 Router D Router G

The EIGRP Topology Table (continued)

Remember, however, that Router A's Feasible Distance must be the route with the lowest metric. If we add the Advertised Distance with the local metric between each router, we would see that:

- The route through Router B has a distance of **16** to the destination
- The route through Router C has a distance of 27 to the destination
- The route through Router D has a distance of 11 to the destination

Thus, the route through Router D (metric of 11) would become the **Feasible Distance** for Router A, and is added to the routing table as the *best* route. This route is identified as the **Successor**.

To allow convergence to occur quickly if a link fails, EIGRP includes backup routes in the topology table called **Feasible Successors** (**FS**). A route will only become a Successor if its Advertised Distance is *less* than the current Feasible Distance. This is known as a **Feasible Condition** (**FC**).

For example, we determined that Router A's Feasible Distance to the destination is 11, through Router D. Router C's Advertised Distance is 23, and thus would not become a Feasible Successor, as it has a higher metric than Router A's current Feasible Distance. Routes that are not Feasible Successors become route Possibilities.

Router B's Advertised Distance is 8, which is less than Router A's current Feasible Distance. Thus, the route through Router B to the Destination Network *would* become a Feasible Successor.

Feasible Successors provide EIGRP with redundancy, without forcing routers to re-converge (thus stopping the flow of traffic) when a topology change occurs. If no Feasible Successor exists and a link fails, a route will enter an **Active** (converging) state until an alternate route is found.

EIGRP Packet Types

EIGRP employs five packet types:

- Hello packets multicast
- **Update packets –** *unicast* or *multicast*
- **Query packets** *multicast*
- Reply packets unicast
- Acknowledgement packets unicast

Hello packets are used to form neighbor relationships, and were explained in detail previously. Hello packets are always multicast to address 224.0.0.10.

Update packets are sent between neighbors to build the topology and routing tables. Updates sent to *new* neighbors are sent as unicasts. However, if a route's metric is changed, the update is sent out as a multicast to address 224.0.0.10.

Query packets are sent by a router when a Successor route fails, and there are no Feasible Successors in the topology table. The router places the route in an **Active state**, and queries its neighbors for an alternative route. Query packets are sent as a multicast to address 224.0.0.10.

Reply packets are sent in response to Query packets, assuming the responding router has an alternative route (feasible successor). Reply packets are sent as a unicast to the querying router.

Recall that EIGRP utilizes the **Reliable Transport Protocol** (**RTP**) to ensure reliable delivery of most EIGRP packets. Delivery is guaranteed by having packets acknowledged using.....Acknowledgment packets!

Acknowledgment packets (also known as **ACK's**) are simply Hello packets with no data, other than an acknowledgment number. ACK's are always sent as unicasts. The following packet types employ RTP to ensure reliable delivery via ACK's:

- Update Packets
- Query Packets
- Reply Packets

Hello and Acknowledgments (ha!) packets do not utilize RTP, and thus do not require acknowledgement.

EIGRP Route States

An EIGRP route can exist in one of two states, in the topology table:

- Active state
- Passive State

A **Passive state** indicates that a route is reachable, and that EIGRP is fully converged. A stable EIGRP network will have all routes in a Passive state.

A route is placed in an **Active state** when the Successor and any Feasible Successors fail, forcing the EIGRP to send out Query packets and reconverge. Multiple routes in an Active state indicate an unstable EIGRP network. If a Feasible Successor exists, a route should *never* enter an Active state.

Routes will become **Stuck-in-Active** (**SIA**) when a router sends out a Query packet, but does not receive a Reply packet within three minutes. In other words, a route will become SIA if EIGRP fails to re-converge.

To view the current state of routes in the EIGRP topology table:

Router# *show ip eigrp topology*

```
IP-EIGRP Topology Table for AS(10)/ID(172.19.1.1)
Codes: P - Passive, A - Active, U - Update, Q - Query, R - Reply,
           r - reply Status, s - sia Status
P 10.3.0.0/16, 1 successors, FD is 2297856
        via 172.16.1.2 (2297856/128256), Serial0
P 172.19.0.0/16, 1 successors, FD is 281600
        via Connected, Serial 1
```

To view only *active* routes in the topology table:

Router# show ip eigrp topology active

```
IP-EIGRP Topology Table for AS(10)/ID(172.19.1.1)
Codes: P - Passive, A - Active, U - Update, Q - Query, R - Reply,
           r - Reply status
A 172.19.0.0/16, 1 successors, FD is 23456056 1 replies,
            active 0:00:38, query-origin: Multiple Origins
```

(Reference: http://www.cisco.com/en/US/tech/tk365/technologies_tech_note09186a008010f016.shtml)

* * *

EIGRP Metrics

EIGRP can utilize 5 separate metrics to determine the best route to a destination:

- **Bandwidth** (K1) Slowest link in the route path, measured in kilobits
- Load (K2) Cumulative load of all outgoing interfaces in the path, given as a fraction of 255
- **Delay of the Line** (K3) Cumulative delay of all outgoing interfaces in the path in tens of microseconds
- Reliability (K4) Average reliability of all outgoing interfaces in the path, given as a fraction of 255
- MTU (K5) The smallest Maximum Transmission Unit in the path. The MTU value is actually *never* used to calculate the metric

By default, only **Bandwidth** and **Delay of the Line** are used. This is identical to IGRP, except that EIGRP provides a more granular metric by multiplying the bandwidth and delay by 256. Bandwidth and delay are determined by the interfaces that lead towards the destination network.

By default, the full formula for determining the EIGRP metric is:

$$[10000000/bandwidth + delay] * 256$$

The bandwidth value represents the link with the *lowest* bandwidth in the path, in kilobits. The delay is the total delay of all outgoing interfaces in the path.

As indicated above, each metric is symbolized with a "K" and then a number. When configuring EIGRP metrics, we actually identify which metrics we want EIGRP to consider. Again, by default, only Bandwidth and Delay are considered. Thus, using on/off logic:

$$K1 = 1$$
, $K2 = 0$, $K3 = 1$, $K4 = 0$, $K5 = 0$

If all metrics were set to "on," the full formula for determining the EIGRP metric would be:

Remember, the "K" value is either set to on ("1") or off ("0").

Configuring EIGRP Metrics

EIGRP allows us to identify which metrics the protocol should consider, using the following commands:

```
Router(config)# router eigrp 10
Router(config-router)# metric weights 0 1 1 1 0 0
```

The first command enables the *EIGRP* process for Autonomous System 10. The second actually identifies which EIGRP metrics to use. The first number (0) is for Type of Service, and should always be zero. The next numbers, in order, are K1 (1), K2 (1), K3 (1), K4 (0), and K5 (0). Thus, we are instructing EIGRP to use bandwidth, load, and delay to calculate the total metric, but not reliability or MTU.

Our formula would thus be:

```
[K1 * bandwidth * 256 + (K2 * bandwidth) / (256 - load)
                 + K3 * delay * 256]
```

The actual values of our metrics (such as bandwidth, delay, etc.) must be configured indirectly. We can adjust the bandwidth of an interface:

```
Router(config)# int s0/0
Router(config-if)# bandwidth 64000
Router(config-if)# ip bandwidth-percent eigrp 10 30
```

However, this command does not actually dictate the *physical* speed of the interface. It merely controls how EIGRP considers this interface. Best practice is to set the bandwidth to the actual physical speed of the interface.

The *ip bandwidth-percent eigrp* command limits the percentage of bandwidth EIGRP can use on an interface. The percentage is based on the configured bandwidth value. By default, EIGRP will use **up to 50%** of the bandwidth of an interface. The above command adjusts this to 30% for Autonomous System 10.

If adjustments to the EIGRP metric need to be made, the delay metric (in tens of microseconds) on an interface should be used:

```
Router(config)# int s0/0
Router(config-if)# delay 10000
```

Metric settings must be **identical** on the connecting interfaces of two routers; otherwise they will not form a neighbor relationship.

Configuring Basic EIGRP



Routing protocol configuration occurs in Global Configuration mode. On Router A, to configure EIGRP, we would type:

> RouterA(config)# router eigrp 10 RouterA(config-router)# network 172.16.0.0 RouterA(config-router)# network 10.0.0.0

The first command, router eigrp 10, enables the EIGRP process. The "10" indicates the Autonomous System number that we are using. The Autonomous System number can range from 1 to 65535.

Only other EIGRP routers in Autonomous System 10 will form neighbor adjacencies and share updates with this router.

The *network* statements serve two purposes in EIGRP:

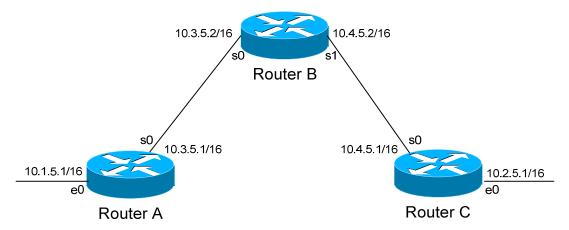
- First, they identify which *networks* you wish to advertise to other EIGRP routers (similar to RIP).
- Second, they identify which *interfaces* on the local router to attempt to form neighbor relationships out of (similar to OSPF).

Prior to IOS version 12.0(4), the *network* statements were classful, despite the fact that EIGRP is a classless routing protocol. For example, the above network 10.0.0.0 command would advertise the networks of directlyconnected interfaces belonging to the 10.0.0.0/8 network and its subnets. It would further attempt to form neighbor relationships out of these interfaces.

IOS version 12.0(4) and later provided us with more granular control of our *network* statements. It introduced a *wildcard mask* parameter, which allows us to choose the networks to advertise in a classless fashion:

> **Router**A(config)# router eigrp 10 **RouterA(config-router)**# network 172.16.0.0 0.0.255.255 RouterA(config-router)# network 10.1.4.0 0.0.0.255

EIGRP Passive Interfaces



It is possible to control which router interfaces will participate in the EIGRP process. Just as with RIP, we can use the *passive-interface* command.

However, please note that the *passive-interface* command works differently with EIGRP than with RIP or IGRP. EIGRP will no longer form neighbor relationships out of a "passive" interface, thus this command prevents updates from being *sent* or *received* out of this interface:

> RouterC(config)# router eigrp 10 RouterC(config-router)# network 10.4.0.0 **RouterC**(config-router)# network 10.2.0.0 **RouterC(config-router)**# passive-interface s0

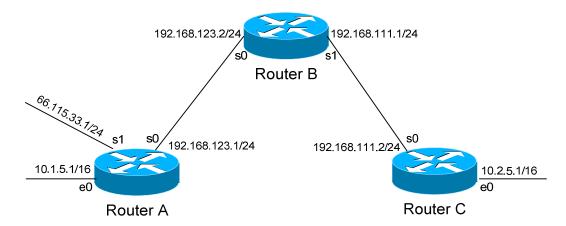
Router C will not form a neighbor adjacency with Router B.

We can configure **all** interfaces to be passive using the *passive-interface* default command, and then individually use the no passive-interface command on the interfaces we **do** want neighbors to be formed on:

> RouterC(config)# router eigrp 10 RouterC(config-router)# network 10.4.0.0 RouterC(config-router)# network 10.2.0.0 **RouterC(config-router)**# passive-interface default **RouterC(config-router)**# no passive-interface e0

Always remember, that the *passive-interface* command will prevent EIGRP (and OSPF) from forming neighbor relationships out of that interface. No routing updates are passed in either direction.

EIGRP Auto-Summarization



EIGRP is a classless routing protocol that supports Variable Length Subnet Masks (VLSMs). The above example would pose no problem for EIGRP.

However, EIGRP will still automatically summarize when crossing major network boundaries.

For example, when Router A sends an EIGRP update to Router B via Serial0, by default it will still summarize the 10.1.0.0/16 network to 10.0.0.0/8. This is because the 10.1.0.0/16 and 192.168.123.0/24 networks do not belong to the same major network. Likewise, the 66.115.33.0/24 network will be summarized to 66.0.0.0/8.

An auto-summary route will be advertised as a normal internal EIGRP route. The *best* metric from among the summarized routes will be applied to this summary route.

The router that *performed* the auto-summarization will also add the summary route to its routing table, with a next hop of the Null0 interface. This is to prevent routing loops.

This **auto-summarization** can be disabled:

RouterA(config)# router eigrp 10 **RouterA(config-router)**# *no auto-summary*

The *no auto-summary* command will prevent Router A from summarizing the 10.1.0.0/16 and 66.115.33.0/24 networks.

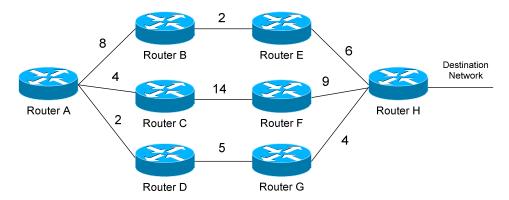
(Reference: http://www.cisco.com/en/US/tech/tk365/technologies_white_paper09186a0080094cb7.shtml#summarization)

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EIGRP Load-Balancing

By default, EIGRP will automatically load-balance across equal-metric routes (four by default, six maximum). EIGRP also supports load-balancing across routes with an unequal metric.

Consider the following example:



Earlier in this section, we established that Router A would choose the route through Router D as its **Feasible Distance** to the destination network. The route through Router B became a Feasible Successor.

By default, EIGRP will *not* load-balance between these two routes, as their metrics are different (11 through Router D, 16 through Router B). We must use the variance command to tell EIGRP to load-balance across these unequal-metric links:

> **RouterA(config)#** router eigrp 10 **RouterA(config-router)**# variance 2 **RouterA**(config-router)# maximum-paths 6

The variance command assigns a "multiplier," in this instance of 2. We multiply this variance value by the metric of our Feasible Distance (2 x 11 = 22). Thus, any Feasible Successors with a metric within twice that of our Feasible Distance (i.e. 12 through 22) will now be used for load balancing by EIGRP.

Remember, only Feasible Successors can be used for load balancing, not Possibilities (such as the route through Router C).

The maximum-paths command adjusts the number of links EIGRP can loadbalance across.

Troubleshooting EIGRP

To view the EIGRP Neighbor Table:

Router# *show ip eigrp neighbor*

IP-EIGRP neighbors for process 10

	Address						_	Seq Num	Type
0	172.16.1.2	S0	13	00:00:53	32	200	0	2	
0	172.18.1.2	S2	11	00:00:59	32	200	0	3	

To view the EIGRP Topology Table, containing all EIGRP route information:

Router# *show ip eigrp topology*

```
IP-EIGRP Topology Table for AS(10)/ID(172.19.1.1)
Codes: P - Passive, A - Active, U - Update, Q - Query, R - Reply,
           r - reply Status, s - sia Status
P 10.3.0.0/16, 1 successors, FD is 2297856
        via 172.16.1.2 (2297856/128256), Serial0
P 172.19.0.0/16, 1 successors, FD is 281600
        via Connected, Serial 1
P 172.18.0.0/16, 1 successors, FD is 128256
         via Connected, Serial 2
P 172.16.0.0/16, 1 successors, FD is 2169856
         via Connected, Serial0
```

To view information on EIGRP traffic sent and received on a router:

Router# show ip eigrp traffic

```
IP-EIGRP Traffic Statistics for process 10
 Hellos sent/received: 685/429
 Updates sent/received: 4/3
 Oueries sent/received: 0/0
 Replies sent/received: 0/0
 Acks sent/received: 1/2
  Input queue high water mark 1, 0 drops
  SIA-Queries sent/received: 0/0
  SIA-Replies sent/received: 0/0
```

Troubleshooting EIGRP (continued)

To view the bandwidth, delay, load, reliability and MTU values of an interface:

Router# show interface s0

```
SerialO is up, line protocol is up
 Hardware is HD64570
  Internet address is 172.16.1.1/16
 MTU 1500 bytes, BW 1544 Kbit, DLY 20000 usec,
     reliability 255/255, txload 1/255, rxload 1/255
<irrelevant output removed>
```

To view information specific to the EIGRP protocol:

Router# *show ip protocols*

```
Routing Protocol is "eigrp 10"
  Outgoing update filter list for all interfaces is not set
  Incoming update filter list for all interfaces is not set
  Default networks flagged in outgoing updates
  Default networks accepted from incoming updates
  EIGRP metric weight K1=1, K2=0, K3=1, K4=0, K5=0
  EIGRP maximum hopcount 100
  EIGRP maximum metric variance 1
  Redistributing: eigrp 10
  Automatic network summarization is not in effect
  Maximum path: 4
  Routing for Networks:
    172.16.0.0
    172.18.0.0
    172.19.0.0
  Routing Information Sources:
    Gateway Distance Last Update (this router) 90 00:26:11 172.16.1.2 90 00:23:49
  Distance: internal 90 external 170
```

This command provides us with information on EIGRP timers, EIGRP metrics, summarization, and the specific networks RIP is advertising.

Troubleshooting EIGRP (continued)

To view the IP routing table:

Router# *show ip route*

```
Gateway of last resort is not set
      172.16.0.0 is directly connected, Serial0
      172.19.0.0 is directly connected, Serial1
      10.3.0.0 [90/2297856] via 172.16.1.2, 00:00:15, Serial0
```

To view a specific route within the IP routing table:

Router# *show ip route 10.3.0.0*

```
Routing entry for 10.3.0.0/16
  Known via "eigrp 10", distance 90, metric 2297856 type internal
  Last update from 172.16.1.2 on Serial 0, 00:00:15 ago
```

To debug EIGRP in realtime:

Router# *debug eigrp neighbors* Router# debug eigrp packet Router# debug eigrp route Router# debug eigrp summary

Section 20 - Open Shortest Path First -

OSPF (Open Shortest Path First)

OSPF is a standardized Link-State routing protocol, designed to scale efficiently to support larger networks.

OSPF adheres to the following Link State characteristics:

- OSPF employs a hierarchical network design using **Areas**.
- OSPF will form **neighbor** relationships with adjacent routers in the same Area.
- Instead of advertising the *distance* to connected networks, OSPF advertises the *status* of directly connected **links** using **Link-State** Advertisements (LSAs).
- OSPF sends updates (LSAs) when there is a change to one of its links, and will only send the change in the update. LSAs are additionally refreshed every 30 minutes.
- OSPF traffic is multicast either to address **224.0.0.5** (all OSPF routers) or **224.0.0.6** (all Designated Routers).
- OSPF uses the **Dijkstra Shortest Path First** algorithm to determine the shortest path.
- OSPF is a classless protocol, and thus supports VLSMs.

Other characteristics of OSPF include:

- OSPF supports only IP routing.
- OSPF routes have an administrative distance is **110**.
- OSPF uses **cost** as its metric, which is computed based on the bandwidth of the link. OSPF has no hop-count limit.

The OSPF process builds and maintains three separate tables:

- A **neighbor table** contains a list of all neighboring routers.
- A topology table contains a list of *all* possible routes to all known networks within an area.
- A **routing table** contains the *best* route for each known network.

OSPF Neighbors

OSPF forms neighbor relationships, called **adjacencies**, with other routers in the same **Area** by exchanging **Hello** packets to multicast address **224.0.0.5**. Only after an adjacency is formed can routers share routing information.

Each OSPF router is identified by a unique **Router ID**. The Router ID can be determined in one of three ways:

- The Router ID can be **manually** specified.
- If not manually specified, the highest IP address configured on any **Loopback interface** on the router will become the Router ID.
- If no loopback interface exists, the highest IP address configured on any **Physical interface** will become the Router ID.

By default, Hello packets are sent out OSPF-enabled interfaces every 10 seconds for broadcast and point-to-point interfaces, and 30 seconds for nonbroadcast and point-to-multipoint interfaces.

OSPF also has a **Dead Interval**, which indicates how long a router will wait without hearing any hellos before announcing a neighbor as "down." Default for the Dead Interval is 40 seconds for broadcast and point-to-point interfaces, and 120 seconds for non-broadcast and point-to-multipoint interfaces. Notice that, by default, the dead interval timer is four times the Hello interval.

These timers can be adjusted on a per interface basis:

Router(config-if)# ip ospf hello-interval 15 Router(config-if)# ip ospf dead-interval 60

OSPF Neighbors (continued)

OSPF routers will only become neighbors if the following parameters within a Hello packet are identical on each router:

- Area ID
- Area Type (stub, NSSA, etc.)
- Prefix
- Subnet Mask
- Hello Interval
- Dead Interval
- Network Type (broadcast, point-to-point, etc.)
- Authentication

The Hello packets also serve as **keepalives** to allow routers to quickly discover if a neighbor is down. Hello packets also contain a **neighbor field** that lists the Router IDs of all neighbors the router is connected to.

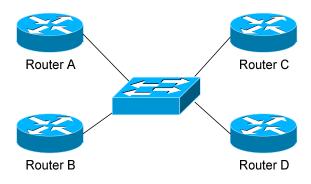
A **neighbor table** is constructed from the OSPF Hello packets, which includes the following information:

- The **Router ID** of each neighboring router
- The current "state" of each neighboring router
- The interface directly connecting to each neighbor
- The IP address of the remote interface of each neighbor

(Reference: http://www.cisco.com/warp/public/104/29.html)

OSPF Designated Routers

In multi-access networks such as Ethernet, there is the possibility of many neighbor relationships on the same physical segment. In the above example, four routers are connected into the same multi-access segment. Using the following formula (where "n" is the number of routers):



n(n-1)/2

....it is apparent that 6 separate adjacencies are needed for a fully meshed network. Increase the number of routers to five, and 10 separate adjacencies would be required. This leads to a considerable amount of unnecessary Link State Advertisement (LSA) traffic.

If a link off of Router A were to fail, it would flood this information to all neighbors. Each neighbor, in turn, would then flood that same information to all other neighbors. This is a waste of bandwidth and processor load.

To prevent this, OSPF will elect a **Designated Router** (**DR**) for each multiaccess networks, accessed via multicast address **224.0.0.6**. For redundancy purposes, a **Backup Designated Router (BDR)** is also elected.

OSPF routers will form adjacencies with the DR and BDR. If a change occurs to a link, the update is forwarded only to the DR, which then forwards it to all other routers. This greatly reduces the flooding of LSAs.

DR and BDR elections are determined by a router's **OSPF priority**, which is configured on a per-interface basis (a router can have interfaces in multiple multi-access networks). The router with the **highest priority** becomes the DR; second highest becomes the BDR. If there is a tie in priority, whichever router has the highest Router ID will become the DR. To change the priority on an interface:

Router(config-if)# ip ospf priority 125

Default priority on Cisco routers is 1. A priority of 0 will prevent the router from being elected DR or BDR. Note: The DR election process is not preemptive. Thus, if a router with a higher priority is added to the network, it will not automatically supplant an existing DR. Thus, a router that should never become the DR should always have its priority set to 0.

OSPF Neighbor States

Neighbor adjacencies will progress through several **states**, including:

Down – indicates that no Hellos have been heard from the neighboring router.

Init – indicates a Hello packet has been heard from the neighbor, but twoway communication has not yet been initialized.

2-Way – indicates that bidirectional communication has been established. Recall that Hello packets contain a *neighbor* field. Thus, communication is considered 2-Way once a router sees its own Router ID in its neighbor's Hello Packet. Designated and Backup Designated Routers are elected at this stage.

ExStart – indicates that the routers are preparing to share link state information. Master/slave relationships are formed between routers to determine who will begin the exchange.

Exchange – indicates that the routers are exchanging Database Descriptors (**DBDs**). DBDs contain a description of the router's Topology Database. A router will examine a neighbor's DBD to determine if it has information to share.

Loading – indicates the routers are finally exchanging **Link State** Advertisements, containing information about all links connected to each router. Essentially, routers are sharing their topology tables with each other.

Full – indicates that the routers are fully synchronized. The topology table of all routers in the area should now be identical. Depending on the "role" of the neighbor, the state may appear as:

- Full/DR indicating that the neighbor is a Designated Router (DR)
- Full/BDR indicating that the neighbor is a Backup Designated Router (BDR)
- Full/DROther indicating that the neighbor is neither the DR or **BDR**

On a multi-access network, OSPF routers will *only* form Full adjacencies with DRs and BDRs. Non-DRs and non-BDRs will still form adjacencies. but will remain in a 2-Way State. This is normal OSPF behavior.

OSPF Network Types

OSPF's functionality is different across several different network topology types. OSPF's interaction with Frame Relay will be explained in another section

Broadcast Multi-Access – indicates a topology where broadcast occurs.

- Examples include Ethernet, Token Ring, and ATM.
- OSPF will elect DRs and BDRs.
- Traffic to DRs and BDRs is multicast to 224,0,0,6. Traffic from DRs and BDRs to other routers is multicast to 224.0.0.5.
- Neighbors *do not* need to be manually specified.

Point-to-Point – indicates a topology where two routers are directly connected.

- An example would be a point-to-point T1.
- OSPF will not elect DRs and BDRs.
- All OSPF traffic is multicast to 224.0.0.5.
- Neighbors *do not* need to be manually specified.

Point-to-Multipoint – indicates a topology where one interface can connect to multiple destinations. Each connection between a source and destination is treated as a point-to-point link.

- An example would be Point-to-Multipoint Frame Relay.
- OSPF will not elect DRs and BDRs.
- All OSPF traffic is multicast to 224.0.0.5.
- Neighbors *do not* need to be manually specified.

Non-broadcast Multi-access Network (NBMA) – indicates a topology where one interface can connect to multiple destinations; however, broadcasts cannot be sent across a NBMA network.

- An example would be Frame Relay.
- OSPF will elect DRs and BDRs.
- OSPF neighbors must be *manually* defined, thus All OSPF traffic is unicast instead of multicast.

Remember: on *non-broadcast* networks, neighbors must be **manually specified**, as multicast Hello's are not allowed.

Configuring OSPF Network Types

The default OSPF network type for basic Frame Relay is Non-broadcast Multi-access Network (NBMA). To configure manually:

> **Router(config)**# interface s0 **Router**(**config-if**)# *encapsulation frame-relay* **Router(config-if)**# frame-relay map ip 10.1.1.1 101 **Router(config-if)**# ip ospf network non-broadcast

Router(config)# router ospf 1 **Router(config-router)**# neighbor 10.1.1.1

Notice that the *neighbor* was manually specified, as multicasting is not allowed on an NBMA. However, the Frame-Relay network can be tricked into allowing broadcasts, eliminating the need to manually specify neighbors:

> **Router(config)#** interface s0 **Router(config-if)**# *encapsulation frame-relay* **Router(config-if)**# frame-relay map ip 10.1.1.1 101 broadcast **Router(config-if)**# ip ospf network broadcast

Notice that the *ospf network* type has been changed to *broadcast*, and the broadcast parameter was added to the frame-relay map command. The neighbor no longer needs to be specified, as multicasts will be allowed out this map.

The default OSPF network type for Ethernet and Token Ring is **Broadcast** Multi-Access. To configure manually:

> **Router(config)#** *interface e0* **Router(config-if)**# ip ospf network broadcast

The default OSPF network type for T1's (HDLC or PPP) and Point-to-Point Frame Relay is **Point-to-Point.** To configure manually:

> **Router(config)#** *interface s0* **Router(config-if)**# encapsulation frame-relay

Router(config)# interface s0.1 point-to-point **Router(config-if)**# frame-relay map ip 10.1.1.1 101 broadcast **Router(config-if)**# ip ospf network point-to-point

Configuring OSPF Network Types (continued)

Router(config)# *interface s0*

The default OSPF network type for Point-to-Multipoint Frame Relay is *still* Non-broadcast Multi-access Network (NBMA). However, OSPF supports an additional network type called **Point-to-Multipoint**, which will allow neighbor discovery to occur automatically. To configure:

```
Router(config-if)# encapsulation frame-relay
Router(config)# interface s0.2 multipoint
Router(config-if)# frame-relay map ip 10.1.1.1 101 broadcast
Router(config-if)# ip ospf network point-to-multipoint
```

Additionally, a non-broadcast parameter can be added to the ip ospf network command when specifying point-to-multipoint.

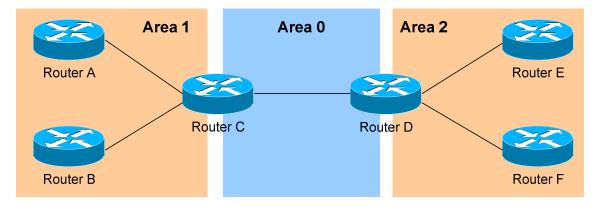
```
Router(config)# interface s0
Router(config-if)# encapsulation frame-relay
Router(config)# interface s0.2 multipoint
Router(config-if)# frame-relay map ip 10.1.1.1 101
Router(config-if)# ip ospf network point-to-multipoint non-broadcast
Router(config)# router ospf 1
Router(config-router)# neighbor 10.1.1.1
```

Notice the different in configuration. The *frame-relay map* command no longer has the *broadcast* parameter, as broadcasts and multicasts are not allowed on a non-broadcast network.

Thus, in the OSPF router configuration, neighbors must again be manually specified. Traffic to those neighbors will be unicast instead of multicast.

OSPF network types must be set identically on two "neighboring" routers, otherwise they will never form an adjacency.

The OSPF Hierarchy



OSPF is a hierarchical system that separates an Autonomous System into individual areas. OSPF traffic can either be intra-area (within one area), **inter-area** (between separate areas), or **external** (from another AS).

OSPF routers build a **Topology Database** of all **links** within their area, and all routers within an area will have an *identical* topology database. Routing updates between these routers will *only* contain information about links local to their area. Limiting the topology database to include only the local area conserves bandwidth and reduces CPU loads.

Area 0 is required for OSPF to function, and is considered the "Backbone" area. As a rule, all other areas must have a connection into Area 0, though this rule can be bypassed using **virtual links** (explained shortly). Area 0 is often referred to as the *transit* area to connect all other areas.

OSPF routers can belong to multiple areas, and will thus contain separate Topology databases for each area. These routers are known as **Area Border** Routers (ABRs).

Consider the above example. Three areas exist: Area 0, Area 1, and Area 2. Area 0, again, is the backbone area for this Autonomous System. Both Area 1 and Area 2 must directly connect to Area 0.

Routers A and B belong fully to Area 1, while Routers E and F belong fully to Area 2. These are known as **Internal Routers**.

Router C belongs to both Area 0 and Area 1. Thus, it is an **ABR**. Because it has an interface in Area 0, it can also be considered a **Backbone Router**. The same can be said for Router D, as it belongs to both Area 0 and Area 2.

The OSPF Hierarchy (continued) External Networks Area 1 Area 2 Router G Router E Router A Router C Router D Area 0 Router B Router F

Now consider the above example. Router G has been added, which belongs to Area 0. However, Router G also has a connection to the Internet, which is outside this Autonomous System.

This makes Router G an Autonomous System Border Router (ASBR). A router can become an ASBR in one of two ways:

- By connecting to a separate Autonomous System, such as the Internet
- By redistributing another routing protocol into the OSPF process.

ASBRs provide access to external networks. OSPF defines two "types" of external routes:

- Type 2 (E2) Includes only the external cost to the destination network. External cost is the metric being advertised from outside the OSPF domain. This is the default type assigned to external routes.
- Type 1 (E1) Includes both the external cost, and the internal cost to reach the ASBR, to determine the total metric to reach the destination network. Type 1 routes are always *preferred* over Type 2 routes to the same destination.

Thus, the four separate OSPF router types are as follows:

- **Internal Routers** all router interfaces belong to only one Area.
- Area Border Routers (ABRs) contains interfaces in at least two separate areas
- **Backbone Routers** contain at least one interface in Area 0
- Autonomous System Border Routers (ASBRs) contain a connection to a separate Autonomous System

LSAs and the OSPF Topology Database

OSPF, as a link-state routing protocol, does not rely on routing-by-rumor as RIP and IGRP do.

Instead, OSPF routers keep track of the status of **links** within their respective areas. A link is simply a router interface. From these lists of links and their respective statuses, the topology database is created. OSPF routers forward link-state advertisements (LSAs) to ensure the topology database is consistent on each router within an area.

Several LSA types exist:

- Router LSA (Type 1) Contains a list of all links local to the router, and the status and "cost" of those links. Type 1 LSAs are generated by all routers in OSPF, and are flooded to all other routers within the local area.
- Network LSA (Type 2) Generated by all Designated Routers in OSPF, and contains a list of all routers attached to the Designated Router.
- Network Summary LSA (Type 3) Generated by all ABRs in OSPF, and contains a list of all destination networks within an area. Type 3 LSAs are sent between areas to allow inter-area communication to occur.
- ASBR Summary LSA (Type 4) Generated by ABRs in OSPF, and contains a *route* to any ASBRs in the OSPF system. Type 4 LSAs are sent from an ABR into its local area, so that Internal routers know how to exit the Autonomous System.
- External LSA (Type 5) Generated by ASBRs in OSPF, and contain routes to destination networks *outside* the local Autonomous System. Type 5 LSAs can also take the form of a **default route** to all networks outside the local AS. Type 5 LSAs are flooded to all areas in the OSPF system.

Multicast OSPF (MOSPF) utilizes a Type 6 LSA, but that goes beyond the scope of this guide.

Later in this section, Type 7 NSSA External LSAs will be described in detail.

External Networks Area 1 Area 2 S0/0 Router G Router A Router E Fa0/0 Fa0/0 Fa0/1 Router D Router C Fa0/0 Fa0/0 Router B Area 0 Router F

LSAs and the OSPF Topology Database (continued)

From the above example, the following can be determined:

- Routers A, B, E, and F are **Internal Routers**.
- Routers C and D are **ABRs**.
- Router G is an ASBR.

All routers will generate **Router** (**Type 1**) **LSAs**. For example, Router A will generate a Type 1 LSA that contains the status of links FastEthernet 0/0 and FastEthernet 0/1. This LSA will be flooded to all other routers in Area 1.

Designated Routers will generate **Network** (**Type 2**) **LSAs.** For example, if Router C was elected the DR for the multi-access network in Area 1, it would generate a Type 2 LSA containing a list of all routers attached to it.

Area Border Routers (ABRs) will generate **Network Summary** (**Type 3**) LSAs. For example, Router C is an ABR between Area 0 and Area 1. It will thus send Type 3 LSAs into both areas. Type 3 LSAs sent into Area 0 will contain a list of networks within Area 1, including costs to reach those networks. Type 3 LSAs sent into Area 1 will contain a list of networks within Area 0, and all other areas connected to Area 0. This allows Area 1 to reach any other area, and all other areas to reach Area 1.

External Networks Area 1 Area 2 S0/0 Router G Router A Router E Fa0/0 Fa0/0 Fa0/1 Router D Router C Fa0/0 Router B Area 0 Router F

LSAs and the OSPF Topology Database (continued)

ABRs will also generate **ASBR Summary** (**Type 4**) **LSAs.** For example, Router C will send Type 4 LSAs into Area 1 containing a route to the ASBR, thus providing routers in Area 1 with the path out of the Autonomous System.

ASBRs will generate External (Type 5) LSAs. For example, Router G will generate Type 5 LSAs that contain routes to network outside the AS. These Type 5 LSAs will be flooded to routers of all areas.

Each type of LSA is propagated under three circumstances:

- When a new adjacency is formed.
- When a change occurs to the topology table.
- When an LSA reaches its maximum age (every **30 minutes,** by default).

Thus, though OSPF is typically recognized to only send updates when a change occurs, LSA's are still periodically refreshed every 30 minutes.

The OSPF Metric

OSPF determines the best (or *shortest*) path to a destination network using a **cost** metric, which is based on the bandwidth of interfaces. The *total* cost of a route is the sum of all *outgoing* interface costs. Lowest cost is preferred.

Cisco applies default costs to specific interface types:

Type	Cost
Serial (56K)	1785
Serial (64K)	1562
<i>T1 (1.544Mbps)</i>	64
Token Ring (4Mbps)	25
Ethernet (10 Mbps)	10
Token Ring (16 Mbps)	6
Fast Ethernet	1

On Serial interfaces, OSPF will use the configured *bandwidth* (measured in Kbps) to determine the cost:

> **Router(config)#** *interface s0* Router(config-if)# bandwidth 64

The default cost of an interface can be superseded:

Router(config)# interface e0 **Router(config-if)**# ip ospf cost 5

Changing the cost of an interface can alter which path OSPF deems the "shortest," and thus should be used with great care.

To alter how OSPF calculates its default metrics for interfaces:

Router(config)# router ospf 1 **Router(config-router)**# ospf auto-cost reference-bandwidth 100

The above *ospf auto-cost* command has a value of 100 configured, which is actually the default. This indicates that a 100Mbps link will have a cost of 1 (because 100/100 is 1). All other costs are based off of this. For example, the cost of 4 Mbps Token Ring is 25 because 100/4 = 25.

Configuring Basic OSPF



Routing protocol configuration occurs in Global Configuration mode. On Router A, to configure OSPF:

> **RouterA**(config)# router ospf 1 **RouterA**(config-router)# router-id 1.1.1.1 **RouterA(config-router)**# network 172.16.0.0 0.0.255.255 area 1 **RouterA(config-router)**# network 172.17.0.0 0.0.255.255 area 0

The first command, router ospf 1, enables the OSPF process. The "1" indicates the OSPF process ID, and can be unique on each router. The process ID allows multiple OSPF processes to run on the same router. The router-id command assigns a unique OSPF ID of 1.1.1.1 for this router.

Note the use of a wildcard mask instead of a subnet mask in the *network* statement. With OSPF, we're *not* telling the router what networks to advertise; we're telling the router to place certain interfaces into specific areas, so those routers can form neighbor relationships. The wildcard mask 0.0.255.255 tells us that the last two octets can match any number.

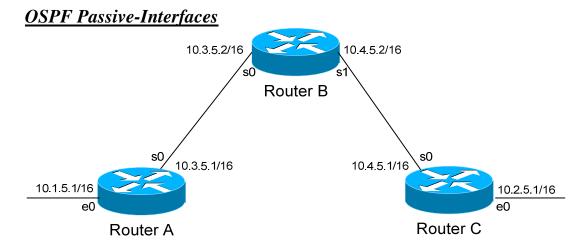
The first *network* statement places interface E0 on Router A into Area 1. Likewise, the second *network* statement places interface S0 on Router A into Area 0. The network statement could have been written more specifically:

> **RouterA(config)#** router ospf 1 RouterA(config-router)# network 172.16.1.2 0.0.0.0 area 1 RouterA(config-router)# network 172.17.1.1 0.0.0.0 area 0

In order for Router B to form a neighbor relationship with Router A, its connecting interface must be put in the same Area as Router A:

> **RouterB(config)**# router ospf 1 **RouterA(config-router)#** router-id 2.2.2.2 RouterB(config-router)# network 172.17.1.2 0.0.0.0 area 0 RouterB(config-router)# network 172.18.1.1 0.0.0.0 area 2

If Router B's S0 interface was placed in a different area than Router A's S0 interface, the two routers would never form a neighbor relationship, and never share routing updates.



It is possible to control which router interfaces will participate in the OSPF process. Just as with EIGRP and RIP, we can use the passive-interface command.

However, please note that the *passive-interface* command works differently with OSPF than with RIP or IGRP. OSPF will no longer form neighbor relationships out of a "passive" interface, thus this command prevents updates from being *sent* or *received* out of this interface:

> **RouterC(config)**# router ospf 1 RouterC(config-router)# network 10.4.0.0 0.0.255.255 area 0 RouterC(config-router)# network 10.2.0.0 0.0.255.255 area 0 **RouterC(config-router)**# passive-interface s0

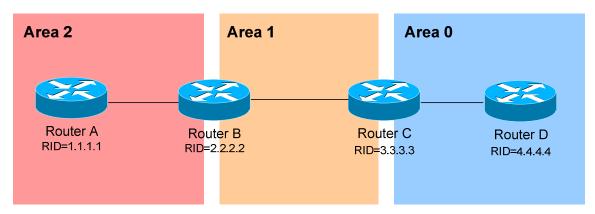
Router C will not form a neighbor adjacency with Router B.

It is possible to configure **all** interfaces to be passive using the passiveinterface default command, and then individually use the no passive*interface* command on the interfaces that neighbors **should** be formed on:

> **RouterC(config)#** router ospf 1 RouterC(config-router)# network 10.4.0.0 0.0.255.255 area 0 RouterC(config-router)# network 10.2.0.0 0.0.255.255 area 0 **RouterC(config-router)**# passive-interface default **RouterC(config-router)**# no passive-interface e0

Always remember, that the *passive-interface* command will prevent OSPF (and EIGRP) from forming neighbor relationships out of that interface. No routing updates are passed in either direction.

OSPF Virtual Links



Earlier in this guide, it was stated that all areas must directly connect into Area 0, as a rule. In the above example, Area 2 has no direct connection to Area 0, but must transit through Area 1 to reach the backbone area. In normal OSPF operation, this shouldn't be possible.

There may be certain circumstances that may prevent an area from directly connecting into Area 0. Virtual links can be used as a workaround, to logically connect separated areas to Area 0. In the above example, a virtual link would essentially create a tunnel from Area 2 to Area 0, using Area 1 a **transit area**. One end of the Virtual Link *must* be connected to Area 0.

Configuration occurs on the **Area Border Routers** (ABRs) connecting Area 1 to Area 2 (Router B), and Area 1 to Area 0 (Router C). Configuration on Router B would be as follows:

```
RouterB(config)# router ospf 1
RouterB(config-router)# router-id 2.2.2.2
RouterB(config-router)# area 1 virtual-link 3.3.3.3
```

The first command enables the *ospf* process. The second command manually sets the *router-id* for Router B to 2.2.2.2.

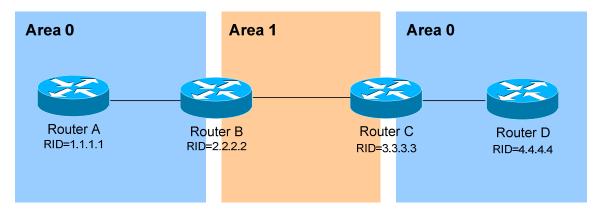
The third command actually creates the *virtual-link*. Notice that it specifies area 1, which is the **transit area**. Finally, the command points to the remote ABR's Router ID of 3.3.3.3.

Configuration on Router C would be as follows:

```
RouterC(config)# router ospf 1
RouterC(config-router)# router-id 3.3.3.3
RouterC(config-router)# area 1 virtual-link 2.2.2.2
```

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OSPF Virtual Links (continued)



It is also possible to have two separated (or discontiguous) Area 0's. In order for OSPF to function properly, the two Area 0's must be connected using a virtual link.

Again, configuration occurs on the transit area's ABRs:

RouterB(config)# router ospf 1

RouterB(config-router)# router-id 2.2.2.2

RouterB(config-router)# area 1 virtual-link 3.3.3.3

RouterC(config)# router ospf 1

RouterC(config-router)# router-id 3.3.3.3

RouterC(config-router)# area 1 virtual-link 2.2.2.2

Always remember: the area specified in the *virtual-link* command is the **transit** area. Additionally, the transit area *cannot* be a stub area.

As stated earlier, if authentication is enabled for Area 0, the same authentication must be configured on Virtual Links, as they are "extensions" of Area 0:

RouterB(config)# router ospf 1

RouterB(config-router)# area 1 virtual-link 3.3.3.3 message-digest-key 1 md5 MYKEY

RouterC(config)# router ospf 1

RouterC(config-router)# area 1 virtual-link 2.2.2.2 message-digest-key 1 md5 MYKEY

Troubleshooting OSPF

To view the OSPF Neighbor Table:

Router# *show ip ospf neighbor*

Neighbor ID	Pri	State	Dead Time	Address	Interface
7.7.7.7	1	FULL/ -	00:00:36	150.50.17.2	Serial0
6.6.6.6	1	FULL/DR	00:00:11	150.50.18.1	Ethernet0

The Neighbor Table provides the following information about each neighbor:

- The *Router ID* of the remote neighbor.
- The OSPF *priority* of the remote neighbor (used for DR/BDR elections).
- The current neighbor *state*.
- The *dead interval* timer.
- The connecting *IP address* of the remote neighbor.
- The local *interface* connecting to the remote neighbor.

To view the OSPF topology table:

Router# show ip ospf database

OSPF Router with ID (9.9.9.9) (Process ID 10)

Router Link States (Area 0)

Link ID	ADV Router	Age	Seq#	Checksum	Link count
7.7.7.7	7.7.7.7	329	0x80000007	0x42A0	2
8.8.8.8	8.8.8.8	291	0x80000007	0x9FFC	1

Summary Net Link States (Area 0)

Link ID	ADV Router	Age	Seq#	Checksum
192.168.12.0	7.7.7.7	103	0x80000005	0x13E4
192.168.34.0	7.7.7.7	105	0x80000003	0x345A

The Topology Table provides the following information:

- The actual *link* (or *route*).
- The *advertising* Router ID.
- The link-state *age* timer.
- The *sequence number* and *checksum* for each entry.

(Reference: http://www.cisco.com/en/US/products/sw/iosswrel/ps5187/products_command_reference_chapter09186a008017d02e.html)

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Troubleshooting OSPF (continued)

To view the specific information about an OSPF process:

Router# *show ip ospf 1*

Routing Process "ospf 1" with ID 9.9.9.9

Supports only single TOS(TOS0) routes

Supports opaque LSA

SPF schedule delay 5 secs, Hold time between two SPFs 10 secs

Minimum LSA interval 5 secs. Minimum LSA arrival 1 secs

Number of external LSA 0. Checksum Sum 0x0

Number of opaque AS LSA 0. Checksum Sum 0x0

Number of DCbitless external and opaque AS LSA 0

Number of DoNotAge external and opaque AS LSA 0

Number of areas in this router is 1. 1 normal 0 stub 0 nssa

External flood list length 0

Area BACKBONE(0)

Number of interfaces in this area is 1

Area has no authentication

SPF algorithm executed 3 times

Area ranges are

Number of LSA 2. Checksum Sum 0xDDEC

Number of opaque link LSA 0. Checksum Sum 0x0

Number of DCbitless LSA 0

Number of indication LSA 0

Number of DoNotAge LSA 0

Flood list length 0

The *show ip ospf* command provides the following information:

- The local *Router ID*.
- *SPF Scheduling* information, and various *SPF timers*.
- The number of *interfaces* in specific *areas*, including the *type* of area.
- The link-state *age* timer.
- The *sequence number* and *checksum* for each entry.

Troubleshooting OSPF (continued)

To view OSPF-specific information on an interface:

```
Router# show ip ospf interface s0
```

Serial0 is up, line protocol is up
Internet Address 192.168.79.2/24, Area 0
Process ID 10, Router ID 9.9.9.9, Network Type POINT_TO_POINT, Cost: 64
Transmit Delay is 1 sec, State POINT_TO_POINT,
Timer intervals configured, Hello 10, Dead 40, Wait 40, Retransmit 5
Hello due in 00:00:04
Index 1/1, flood queue length 0
Next 0x0(0)/0x0(0)
Last flood scan length is 1, maximum is 1
Last flood scan time is 0 msec, maximum is 0 msec
Neighbor Count is 1, Adjacent neighbor count is 1
Adjacent with neighbor 7.7.7.7
Suppress hello for 0 neighbor(s)

The *show ip ospf interface* command provides the following information:

- The local *Router ID*.
- The interface *network type*.
- The OSPF *cost* for the interface.
- The interface *Hello* and *Dead* timers.
- A list of neighbor adjacencies.

To view routing protocol specific information for OSPF:

Router# *show ip protocols*

```
Routing Protocol is "ospf 10"

Invalid after 0 seconds, hold down 0, flushed after 0
Outgoing update filter list for all interfaces is
Incoming update filter list for all interfaces is
Routing for Networks:

192.168.79.0 0.0.0.255 area 0
192.168.109.0 0.0.0.255 area 0
Routing Information Sources:
Gateway Distance Last Update
7.7.7.7 110 00:01:05
Distance: (default is 110)
```

The *show ip protocols* command provides the following information:

- Locally originated *networks* that are being advertised.
- Neighboring *sources* for routing information
- The *administrative distance* of neighboring sources.

* * *

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Troubleshooting OSPF (continued)

To reset an OSPF process, including neighbor adjacencies:

Router# clear ip ospf process

To display information about OSPF virtual-links:

Router# show ip ospf virtual-links

To display routes to both ABRs and ASBRs:

Router# show ip ospf border-routers

To debug OSPF in realtime:

Router# debug ip ospf adj Router# debug ip ospf events Router# debug ip ospf hello

Part IV

VLANs, Access-Lists, and Services

Section 21 - VLANs and VTP -

Review of Collision vs. Broadcast Domains

In a previous guide, it was explained that a "collision domain" is a segment where a collision can occur, and that a Layer-2 switch running in Full Duplex breaks up collision domains. Thus, Layer-2 switches create more collision domains, which results in fewer collisions.

However, Layer-2 switches do not break up broadcast domains, and thus belong to only **one broadcast domain**. Layer-2 switches will forward a broadcast or multicast out every port, excluding the port the broadcast or multicast originated from.

Only Layer-3 devices can break apart broadcast domains. Because of this, Layer-2 switches are not well suited for large, scalable networks. Layer-2 switches make forwarding decisions solely based on Data-Link layer MAC addresses, and thus have no way of differentiating between one network and another.

Virtual LANs (VLANs)

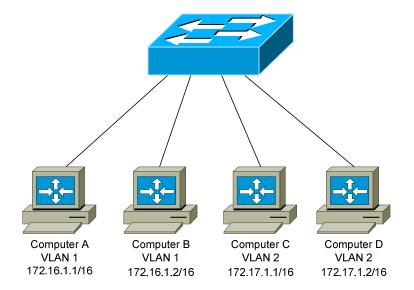
Virtual LANs (or VLANs) separate a Layer-2 switch into multiple **broadcast domains**. Each VLAN is its own individual broadcast domain (i.e. IP subnet).

Individual ports or groups of ports can be assigned to a specific VLAN. Only ports belonging to the same VLAN can freely communicate; ports assigned to separate VLANs require a router to communicate. Broadcasts from one VLAN will never be sent out ports belonging to another VLAN.

Please note: a Layer-2 switch that supports VLANs is not necessarily a Layer-3 switch. A Layer-3 switch, in addition to supporting VLANs, must also be capable of routing, and caching IP traffic flows. Layer-3 switches allow IP packets to be **switched** as opposed to **routed**, which reduces latency.

VLAN Example

Consider the following example:



Four computers are connected to a Layer-2 switch that supports VLANs. Computers A and B belong to VLAN 1, and Computers C and D belong to VLAN 2.

Because Computers A and B belong to the same VLAN, they belong to the same IP subnet and broadcast domain. They will be able to communicate without the need of a router.

Computers C and D likewise belong to the same VLAN and IP subnet. They also can communicate without a router.

However, Computers A and B will *not* be able to communicate with Computers C and D, as they belong to separate VLANs, and thus separate IP subnets. Broadcasts from VLAN 1 will never go out ports configured for VLAN 2. A router will be necessary for both VLANs to communicate.

Most Catalyst multi-layer switches have integrated or modular routing processors. Otherwise, an external router is required for inter-VLAN communication.

By default on Cisco Catalyst switches, all interfaces belong to VLAN 1. VLAN 1 is considered the **Management VLAN** (by default).

Advantages of VLANs

VLANs provide the following advantages:

Broadcast Control – In a pure Layer-2 environment, broadcasts are received by every host on the switched network. In contrast, each VLAN belongs to its own broadcast domain (or IP subnet); thus broadcast traffic from one VLAN will never reach another VLAN.

Security – VLANs allow administrators to "logically" separate users and departments.

Flexibility and Scalability – VLANs remove the physical boundaries of a network. Users and devices can be added or moved anywhere on the physical network, and yet remain assigned to the same VLAN. Thus, access to resources will never be interrupted.

VLAN Membership

VLAN membership can be configured one of two ways:

- Statically Individual (or groups of) switch-ports must be manually assigned to a VLAN. Any device connecting to that switch-port(s) becomes a member of that VLAN. This is a transparent process – the client device is unaware that it belongs to a specific VLAN.
- **Dynamically** Devices are automatically assigned into a VLAN based on its MAC address. This allows a client device to remain in the same VLAN, regardless of which switch port the device is attached to.

Cisco developed a dynamic VLAN product called the VLAN Membership Policy Server (VMPS). In more sophisticated systems, a user's network account can be used to determine VLAN membership, instead of a device's MAC address.

Catalyst switches that *participate in a VTP domain* (explained shortly) support up to 1005 VLANs. Catalyst switches configured in VTP transparent mode support up to 4094 VLANs.

Static VLAN Configuration

The first step in configuring VLANs is to **create** the VLAN:

```
Switch(config)# vlan 100
Switch(config-vlan)# name MY_VLAN
```

The first command creates VLAN 100, and enters VLAN configuration mode. The second command assigns the name MY VLAN to this VLAN. Naming a VLAN is *not* required.

The list of VLANs is stored in Flash in a database file named **vlan.dat**. However, information concerning which local interfaces are assigned to a specific VLAN is not stored in this file; this information is instead stored in the **startup-config** file of each switch.

Next, an interface (or range of interfaces) must be assigned to this VLAN. The following commands will assign interface fa0/10 into the newly created MY_{VLAN} .

```
Switch(config)# interface fa0/10
Switch(config-if)# switchport mode access
Switch(config-if)# switchport access vlan 100
```

The first command enters interface configuration mode. The second command indicates that this is an access port, as opposed to a trunk port (explained in detail shortly). The third command assigns this access port to VLAN 100. Note that the VLAN number is specified, and not the VLAN name.

To view the list of VLANs, including which ports are assigned to each VLAN:

Switch# show vlan

VLAN	Name	Status	Ports
1	default	active	fa0/1-9,11-24
100	MY_VLAN	active	fa0/10
1002	fddi-default	suspended	
1003	token-ring-default	suspended	
1004	fddinet-default	suspended	
1005	trnet-default	suspended	

VLAN Port "Types"

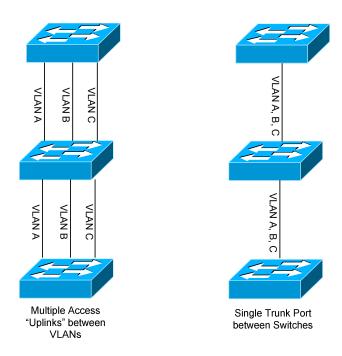
There are two types of ports supported on a VLAN-enabled switch, access ports and trunk ports.

An access port belongs to only one VLAN. Host devices, such as computers and printers, plug into access ports. A host automatically becomes a member of its access port's VLAN. This is done transparently, and the host is usually unaware of the VLAN infrastructure. By default, all switch ports are access ports.

VLANs can span multiple switches. There are two methods of connecting these VLANs together. The first requires creating "uplink" access ports between all switches, for each VLAN. Obviously, in large switching and VLAN environments, this *quickly* becomes unfeasible.

A better alternative is to use **trunk ports**. Trunk ports do not belong to a single VLAN. Any or all VLANs can traverse trunk links to reach other switches. Only Fast or Gigabit Ethernet ports can be used as trunk links.

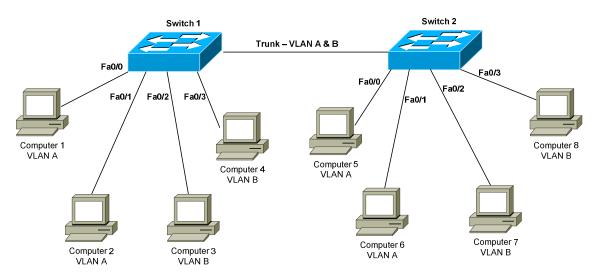
The following diagram illustrates the advantage of using trunk ports, as opposed to uplinking access ports:



VLAN Frame-Tagging

When utilizing trunk links, switches need a mechanism to identify which VLAN a particular frame belongs to. Frame tagging places a VLAN ID in each frame, identifying which VLAN the frame belongs to.

Tagging occurs *only* when a frame is **sent out a trunk port**. Consider the following example:



If Computer 1 sends a frame to Computer 2, no frame tagging will occur. The frame never leaves the Switch 1, stays within its own VLAN, and will simply be **switched** to Computer 2.

If Computer 1 sends a frame to Computer 3, which is in a separate VLAN, frame tagging will still not occur. Again, the frame never leaves the switch, but because Computer 3 is in a different VLAN, the frame must be routed.

If Computer 1 sends a frame to Computer 5, the frame *must* be **tagged** before it is sent out the trunk port. It is stamped with its VLAN ID (in this case, VLAN A), and when Switch 2 receives the frame, it will only forward it out ports belonging to VLAN A (fa0/0, and fa0/1). If Switch 2 has Computer 5's MAC address in its CAM table, it will only send it out the appropriate port (fa0/0).

Cisco switches support two frame-tagging protocols, **Inter-Switch Link** (ISL) and IEEE 802.1Q.

Inter-Switch Link (ISL)

ISL is Cisco's proprietary frame-tagging protocol, and supports Ethernet, Token Ring, FDDI, and ATM frames.

ISL *encapsulates* a frame with an additional header (**26 bytes**) and trailer (**4 bytes**), increasing the size of an Ethernet frame up to **30 bytes**. The header contains the 10 byte VLAN ID. The trailer contains an additional 4-byte CRC for data-integrity purposes.

Because ISL increases the size of a frame, non-ISL devices (i.e. non-Cisco devices) will actually *drop* ISL-tagged frames. Many devices are configured with a maximum acceptable size for Ethernet frames (usually 1514 or 1518 bytes). ISL frames can be as large as 1544 bytes; thus, non-ISL devices will see these packets as **giants** (or corrupted packets).

ISL has deprecated in use over time. Newer Catalyst models may not support ISL tagging.

IEEE 802.1Q

IEEE 802.1Q, otherwise known as DOT1Q, is the standardized frame-tagging protocol supported by most switch manufacturers, including Cisco. Thus, switches from multiple vendors can be trunked together.

Instead of adding an additional header and trailer, 802.1Q actually embeds a **4-byte VLAN ID** into the **Layer-2 frame header**. This still increases the size of a frame from its usual 1514 bytes to 1518 bytes (or from 1518 bytes to 1522 bytes). However, most modern switches support 802.1Q tagging and the slight increase in frame size.

Neither ISL nor 802.1Q tagging alter the source or destination address in the Layer-2 header.

Manual vs. Dynamic Trunking

ISL or 802.1Q tagging can be manually configured on Catalyst trunk ports. Catalyst switches can also dynamically negotiate this using Cisco's proprietary **Dynamic Trunking Protocol (DTP)**.

* * *

Configuring Trunk Links

To manually configure a trunk port, for either ISL or 802.1Q tagging:

Switch(config)# interface fa0/24 Switch(config-if)# switchport trunk encapsulation isl Switch(config-if)# switchport mode trunk

Switch(config)# interface fa0/24 Switch(config-if)# switchport trunk encapsulation dot1q Switch(config-if)# switchport mode trunk

The first line in each set of commands enters interface configuration mode. The second line manually sets the tagging (or *encapsulation*) protocol the trunk link will use. **Always remember**, both sides of the trunk line must be configured with the **same** tagging protocol. The third line manually sets the *switchport mode* to a *trunk* port.

The Catalyst switch can **negotiate** the tagging protocol:

Switch(config)# interface fa0/24 Switch(config-if)# switchport trunk encapsulation negotiate

Whichever tagging protocol is supported on both switches will be used. If the switches support both ISL and 802.1Q, **ISL** will be selected.

By default, trunk ports allow **all VLANs** to traverse the trunk link. However, a list of *allowed* VLANs can be configured on each trunk port:

Switch(config)# interface fa0/24 Switch(config-if)# switchport trunk allowed vlan remove 50-100 Switch(config-if)# switchport trunk allowed vlan add 60-65

The first *switchport* command will prevent the trunk port from passing traffic from VLANs *50-100*. The second *switchport* command will re-allow the trunk port to pass traffic from VLANs *60-65*. In both cases, the *switchport trunk allowed* commands are adding/subtracting from the current list of allowed VLANs, and not replacing that list.

Switch(config)# interface fa0/24 Switch(config-if)# switchport trunk allowed vlan all Switch(config-if)# switchport trunk allowed vlan except 2-99

Certain VLANs are reserved and cannot be removed from a trunk link, including VLAN 1 and system VLANs 1002-1005.

* * *

<u>Native VLANs</u>

A **native VLAN** can also be configured on trunk ports:

Switch(config)# interface fa0/24 **Switch(config-if)**# switchport mode trunk Switch(config-if)# switchport trunk native vlan 42

Frames from the native VLAN are **not tagged** when sent out trunk ports. A trunking interface can only be assigned one native VLAN. Only 802.10 supports native VLANs, whereas ISL does not. (More accurately, ISL will tag frames from all VLANs, even if a VLAN is configured as native). The native VLAN should be configured identically on both sides of the 802.1Q trunk).

Native VLANs are often configured when plugging Cisco VoIP phones into a Catalyst Switch (beyond the scope of this section). Native VLANs are also useful if a **trunk port fails**. For example, if an end user connects a computer into a trunk port, the trunking status will fail and the interface will essentially become an access port. The user's computer will then be transparently *joined* to the Native VLAN.

Native VLANs provide another benefit. A trunk port will accept *untagged* frames and place them in the Native VLAN. Consider the following example:



Assume that both 802.1Q switches have trunk links configured to the non-802.10 switch, and that the trunk ports are configured in Native VLAN 42. Not only will the 802.1Q switches be able to communicate with each other, the non-802.1Q switch will be placed in Native VLAN 42, and be able to communicate with any device in VLAN 42 on any switch.

(Please note, that the author of this study guide finds the "benefit" of the above example of Native VLANs to be.....dubious at best, and confusing as hell at worst).

By default on all trunking interfaces, the Native VLAN is **VLAN 1**.

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Dynamic Trunking Protocol (DTP) Configuration

Not only can the frame tagging protocol of a trunk port be auto-negotiated, but whether a port actually becomes a trunk can be negotiated dynamically as well using the **Dynamic Trunking Protocol (DTP)**.

To manually set a port to be a trunk:

```
Switch(config)# interface fa0/24
Switch(config-if)# switchport mode trunk
```

To allow a port to *dynamically* decide whether to become a trunk, there are two options:

```
Switch(config)# interface fa0/24
Switch(config-if)# switchport mode dynamic desirable
Switch(config)# interface fa0/24
Switch(config-if)# switchport mode dynamic auto
```

If a switchport is set to **dynamic desirable** (the *default* dynamic setting), the interface will actively attempt to form a trunk with the remote switch. If a switchport is set to **dynamic auto**, the interface will passively wait for the remote switch to initiate the trunk.

This results in the following:

- If both ports are manually set to **trunk** a trunk will form.
- If one port is set to **dvnamic desirable**, and the other is set to manual trunk, dynamic desirable, or dynamic auto - a trunk will form.
- If one port is set to **dynamic auto**, and the other port is set to manual trunk or dynamic desirable - a trunk will form.
- If both ports are set to **dynamic auto**, the link will *never* become a trunk, as both ports are waiting for the other to initialize the trunk.

Trunk ports send out DTP frames every 30 seconds to indicate their configured mode.

In general, it is best to manually specific the trunk link, and disable DTP using the *switchport nonegotiate* command:

```
Switch(config)# interface fa0/24
Switch(config-if)# switchport mode trunk
Switch(config-if)# switchport nonegotiate
```

Troubleshooting Trunks

When troubleshooting a misbehaving trunk link, ensure that the following is configured identically on both sides of the trunk:

- **Mode** both sides must be set to *trunk* or *dynamically negotiated*
- Frame-tagging protocol ISL, 802.1Q, or dynamically negotiated
- Native VLAN
- VTP Domain
- Allowed VLANs

If the above parameters are not set identically on both sides, the trunk link will never become active.

To view whether a port is an access or trunk port (such as fa0/5):

Switch# *show interface fa0/24 switchport*

```
Name: Fa0/24
Switchport: Enabled
Administrative Mode: trunk
Operational Mode: trunk
Administrative Trunking Encapsulation: dot1q
Operational Trunking Encapsulation: dot1q
Negotiation of Trunking: On
Access Mode VLAN: 1 (default)
Trunking Native Mode VLAN: 42
<snip>
```

To view the status of all trunk links:

Switch# *show interface trunk*

Port Fa0/24	Mode on	Encapsulation 802.1q	Status trunking	Native VLAN 42
Port Fa0/24	Vlans allowed on trunk 1,100-4094			
Port Fa0/24	Vlans allo 1,100	wed and active in	n managemen	t domain
Port Fa0/24	Vlans in s 1,100	panning tree for	warding sta	te and not pruned

If no interfaces are in a trunking state, the *show interface trunk* command will return no output.

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VLAN Trunking Protocol (VTP)

In large switching environments, it can become difficult to maintain a consistent VLAN database across all switches on the network. The Ciscoproprietary VLAN Trunking Protocol (VTP) allows the VLAN database to be easily managed throughout the network.

Switches configured with VTP are joined to a VTP domain. Only switches belonging to the same domain will share VLAN information, and a switch can only belong to a single domain. When an update is made to the VLAN database, this information is propagated to all switches via VTP advertisements.

By default, VTP updates are sent out every 300 seconds, or anytime a change to the database occurs. VTP updates are sent across VLAN 1, and are only sent out trunk ports.

There are three versions of VTP. The key additions provided by VTP Version 2 are support for Token Ring and Consistency Checks.

VTP Version 1 is default on Catalyst switches, and is **not compatible** with VTP Version 2.

Cisco describes VTP Version 3 as such: "VTP version 3 differs from earlier VTP versions in that it does not directly handle VLANs. VTP version 3 is a protocol that is only responsible for distributing a list of opaque databases over an administrative domain."

(If you are confused, don't be alarmed. The author of this guide is not certain what that means either).

Cisco further defines the enhancements that VTP version 3 provides:

- Support for extended VLANs
- Support for the creation and advertising of private VLANs
- Support for VLAN instances and MST mapping propagation instances
- Improved server authentication
- Protection from the "wrong" database accidently being inserted into a VTP domain.
- Interaction with VTP version 1 and VTP version 2
- Ability to be configured on a per-port basis.

http://www.cisco.com/en/US/docs/switches/lan/catalyst6500/catos/8.x/configuration/guide/vtp.html#wp1017196)

VTP Modes

VTP-enabled switches can operate in one of three **modes**:

- Server
- Client
- Transparent

Only VTP Servers can create, modify or delete entries in the shared VLAN database. Servers advertise their VLAN database to all other switches on the network, including other VTP servers. This is the **default mode** for Cisco Catalyst switches. VTP servers can only advertise VLANs 1 - 1005.

VTP Clients cannot make modifications to the VLAN database, and will receive all of their VLAN information from VTP servers. A client will also forward an update from a server to other clients out its trunk port(s).

Remember, VTP switches must be in the same VTP Domain to share/accept updates to the VLAN database.

A VTP Transparent switch maintains its own separate VLAN database, and will neither advertise nor accept any VLAN database information from other switches (even a server). However, transparent switches will forward VTP updates from servers to clients, thus acting as a pass-through.

Transparent switches handle this pass-through differently depending on the VTP version:

- VTP Version 1 the transparent switch will only pass updates from the same VTP domain.
- VTP Version 2 the transparent switch will pass updates from any VTP domain.

As a best practice, a new switch should be configured as a VTP *client* in the VTP domain, before being installed into a production network. Recall that the default VTP mode on a Cisco Catalyst switch is *server*. If by some circumstance the **configuration revision number** (explained in the next section) is *higher* than that of the existing production switches, a new VTP server could conceivably advertise a blank VLAN database to all other switches.

Configuring the new switch as a VTP client will allow it to *learn* the current VLAN database, and poses no risk to your existing infrastructure.

VTP Updates

VTP updates contain a 32-bit **configuration revision number**, to ensure that all devices have the most current VLAN database. Every change to the VLAN database increments the configuration revision number by 1.

A VTP switch will only accept or synchronize an update if the revision number is *higher* (and thus more recent) than that of the currently installed VLAN database. Updates with a *lower* revision number are ignored.

The simplest way to reset the configuration revision on a VTP server is to change the VTP domain name, and then change it back to the original name.

VTP utilizes three message types:

- Summary Advertisement sent out every 300 seconds, informing all VTP switches of the current configuration revision number.
- **Subset Advertisement** sent out when there is a change to the VLAN database. The subset advertisement actually contains the updated VLAN database.
- Advertisement Request sent out when a switch requires the most current copy of the VLAN database. A switch that is newly joined to the VTP domain will send out an Advertisement Request.

A switch will also send out an Advertisement Request if it receives a Summary Advertisement with a configuration revision number higher than its current VLAN database. A Subset Advertisement will then be sent to that switch, so that it can synchronize the latest VLAN database.

A Subset Advertisement will contain the following fields:

- VTP Version
- VTP Domain
- VTP Configuration Revision
- VLAN IDs for each VLAN in the database
- VLAN-specific information, such as the VLAN name and MTU

(Reference: http://www.cisco.com/en/US/tech/tk389/tk689/technologies_tech_note09186a0080094c52.shtml)

Configuring VTP

To configure the VTP *domain* (the domain name is case sensitive):

Switch(config)# *vtp domain MYDOMAIN*

To configure the VTP *mode*:

```
Switch(config)# vtp mode server
Switch(config)# vtp mode client
Switch(config)# vtp mode transparent
```

The VTP domain can be further secured using a password:

```
Switch(config)# vtp password PASSWORD
```

All switches participating in the VTP domain must be configured with the same password. The password will be hashed into a 16-byte MD5 value.

By default, a Catalyst switch uses VTP version 1. VTP Version 1 and 2 are not compatible. If applied on a VTP server, the following command will enable VTP version 2 globally on all switches:

```
Switch(config)# vtp version 2
```

To view status information about VTP:

Switch# *show vtp status*

```
VTP Version
Configuration Revision : 42
Maximum VLANs supported locally: 1005
Number of existing VLANs : 7
VTP Operating Mode
VTP Domain Name
VTP Pruning Mode
VTP V2 Mode
                                 : Server
                                 : MYDOMAIN
                                 : Disabled
                                 : Enabled
VTP vz mode
VTP Traps Generation
                                 : Disabled
MD5 digest
                                 : 0x42 0x51 0x69 0xBA 0xBE 0xFA 0xCE 0x34
Configuration last modified by 0.0.0.0 at 3-12-09 4:07:52
```

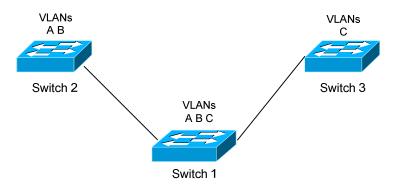
To view VTP statistical information and error counters:

Switch# *show vtp counters*

VTP Pruning

VTP pruning is a process of preventing unnecessary VLAN broadcast or multicast traffic throughout the switching infrastructure.

In the following example, VTP pruning would prevent VLAN C broadcasts from being sent to Switch 2. Pruning would further prevent VLAN A and B broadcast traffic from being sent to Switch 3.



With VTP pruning, traffic is only sent out the necessary VLAN trunk ports where those VLANs exist.

VTP pruning is **disabled by default** on Catalyst IOS switches. If applied on a VTP server, the following command will enable VTP pruning globally on all switches:

Switch(config)# vtp pruning

On trunk ports, it is possible to specify which VLANs are pruning eligible:

Switch(config)# interface fa0/24 **Switch(config-if)**# switchport trunk pruning vlan add 2-50 **Switch(config-if)**# switchport trunk pruning vlan remove 50-100

Switch(config)# interface fa0/24 Switch(config-if)# switchport trunk pruning vlan all **Switch(config-if)**# switchport trunk pruning vlan except 2-100

VLAN 1 is *never* eligible for pruning. The system VLANs 1002-1005 are also pruning-ineligible.

Section 22 - Access Control Lists -

Access Control Lists (ACLs)

Access control lists (ACLs) can be used for two purposes on Cisco devices: to **filter** traffic, and to **identify** traffic.

Access lists are a set of rules, organized in a rule table. Each rule or line in an access-list provides a condition, either **permit** or **deny**:

- When using an access-list to filter traffic, a *permit* statement is used to "allow" traffic, while a *deny* statement is used to "block" traffic.
- Similarly, when using an access list to identify traffic, a *permit* statement is used to "include" traffic, while a deny statement states that the traffic should "not" be included. It is thus interpreted as a true/false statement.

Filtering traffic is the primary use of access lists. However, there are several instances when it is necessary to identify traffic using ACLs, including:

- Identifying interesting traffic to bring up an ISDN link or VPN tunnel
- Identifying routes to filter or allow in routing updates
- Identifying traffic for QoS purposes

When filtering traffic, access lists are applied on interfaces. As a packet passes through a router, the top line of the rule list is checked first, and the router continues to go down the list until a match is made. Once a match is made, the packet is either permitted or denied.

There is an implicit 'deny all' at the end of all access lists. You don't create it, and you can't delete it. Thus, access lists that contain **only deny** statements will prevent all traffic.

Access lists are applied either inbound (packets received on an interface, before routing), or outbound (packets leaving an interface, *after* routing). Only one access list **per interface**, **per protocol**, **per direction** is allowed.

More specific and frequently used rules should be at the top of your access list, to optimize CPU usage. New entries to an access list are added to the bottom. You cannot remove individual lines from a numbered access list. You must delete and recreate the access to truly make changes. Best practice is to use a text editor to manage your access-lists.

Types of Access Lists

There are two categories of access lists: **numbered** and **named**.

Numbered access lists are broken down into several ranges, each dedicated to a specific protocol:

1–99	IP standard access list
100-199	IP extended access list
200-299	Protocol type-code access list
300-399	DECnet access list
400-499	XNS standard access list
500-599	XNS extended access list
600-699	Appletalk access list
700-799	48-bit MAC address access list
800-899	IPX standard access list
900-999	IPX extended access list
1000-1099	IPX SAP access list
1100-1199	Extended 48-bit MAC address access list
1200-1299	IPX summary address access list
1300-1999	IP standard access list (expanded range)
2000-2699	IP extended access list (expanded range

Remember, individual lines *cannot* be removed from a numbered access list. The entire access list must be deleted and recreated. All new entries to a numbered access list are added to the bottom.

Named access lists provide a bit more flexibility. Descriptive names can be used to identify your access-lists. Additionally, individual lines can be removed from a named access-list. However, like numbered lists, all new entries are still added to the bottom of the access list.

There are two common types of named access lists:

- IP standard named access lists
- IP extended named access lists

Configuration of both numbered and named access-lists is covered later in this section.

Wild Card Masks

IP access-lists use **wildcard masks** to determine two things:

- 1. Which part of an address must match exactly
- 2. Which part of an address can match any number

This is as opposed to a **subnet mask**, which tells us what part of an address is the network (subnet), and what part of an address is the host. Wildcard masks look like inversed subnet masks.

Consider the following address and wildcard mask:

Address: 172.16.0.0 Wild Card Mask: 0.0.255.255

The above would match any address that begins "172.16." The last two octets could be anything. How do I know this?

Two Golden Rules of Access Lists:

- 1. If a bit is set to **0** in a wild-card mask, the corresponding bit in the address must be matched exactly.
- 2. If a bit is set to 1 in a wild-card mask, the corresponding bit in the address can match any number. In other words, we "don't care" what number it matches.

To see this more clearly, we'll convert both the address and the wildcard mask into binary:

Address: 10101100.00010000.00000000.00000000 Wild Card Mask: 00000000.000000000.111111111.11111111

Any **0** bits in the wildcard mask, indicates that the corresponding bits in the address must be matched exactly. Thus, looking at the above example, we must exactly match the following in the first two octets:

10101100.00010000 = 172.16

Any 1 bits in the wildcard mask indicates that the corresponding bits can be anything. Thus, the last two octets can be any number, and it will still match this access-list entry.

Wild Card Masks (continued)

If wanted to match a **specific address** with a wildcard mask (we'll use an example of 172.16.1.1), how would we do it?

Address: 172.16.1.1 Wild Card Mask: 0.0.0.0

Written out in binary, that looks like:

Remember what a wildcard mask is doing. A **0** indicates it must match exactly, a **1** indicates it can match anything. The above wildcard mask has all bits set to 0, which means we must match all four octets exactly.

There are actually two ways we can match a host:

- Using a wildcard mask with all bits set to $0 172.16.1.1 \ 0.0.0.0$
- Using the keyword "host" **host 172.16.1.1**

How would we match **all addresses** with a wildcard mask?

Address: 0.0.0.0

Wild Card Mask: 255.255.255.255

Written out in binary, that looks like:

Notice that the above wildcard mask has all bits set to 1. Thus, each bit can match anything – resulting in the above address and wildcard mask matching all possible addresses.

There are actually two ways we can match all addresses:

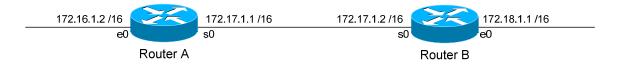
- Using a wildcard mask with all bits set to 1 0.0.0.0255.255.255.255
- Using the keyword "any" any

<u>Standard IP Access List</u>

access-list [1-99] [permit | deny] [source address] [wildcard mask] [log]

Standard IP access-lists are based upon the source host or network IP address, and should be placed closest to the destination network.

Consider the following example:



In order to block network 172.18.0.0 from accessing the 172.16.0.0 network, we would create the following access-list on Router A:

> **Router(config)#** access-list 10 deny 172.18.0.0 0.0.255.255 **Router(config)**# access-list 10 permit any

Notice the wildcard mask of 0.0.255.255 on the first line. This will match (deny) all hosts on the 172.18.x.x network.

The second line uses a keyword of any, which will match (permit) any other address. Remember that you must have at least one permit statement in your access list.

To apply this access list, we would configure the following on Router A:

Router(config)# *int s0* Router(config-if)# ip access-group 10 in

To view all IP access lists configured on the router:

Router# show ip access-list

To view what interface an access-list is configured on:

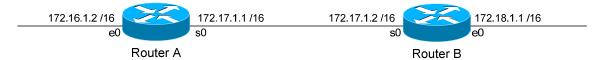
Router# *show ip interface* Router# show running-config

Extended IP Access List

access-list [100-199] [permit | deny] [protocol] [source address] [wildcard mask] [destination address] [wildcard mask] [operator [port]] [log]

Extended IP access-lists block based upon the source IP address, destination IP address, and TCP or UDP port number. Extended access-lists should be placed closest to the source network.

Consider the following example:



Assume there is a webserver on the 172.16.x.x network with an IP address of 172.16.10.10. In order to block network 172.18.0.0 from accessing anything on the 172.16.0.0 network, EXCEPT for the HTTP port on the web server, we would create the following access-list on Router B:

Router(config)# access-list 101 permit tcp 172.18.0.0 0.0.255.255 host 172.16.10.10 eq 80 **Router(config)**# access-list 101 deny ip 172.18.0.0 0.0.255.255 172.16.0.0 0.0.255.255 **Router(config)**# access-list 101 permit ip any any

The first line allows the 172.18.x.x network access only to port 80 on the web server. The second line blocks 172.18.x.x from accessing anything else on the 172.16.x.x network. The third line allows 172.18.x.x access to anything else.

We could have identified the web server in one of two ways:

Router(config)# access-list 101 permit tcp 172.18.0.0 0.0.255.255 host 172.16.10.10 eq 80 **Router(config)**# access-list 101 permit tcp 172.18.0.0 0.0.255.255 172.16.10.10 0.0.0.0 eq 80

To apply this access list, we would configure the following on Router B:

Router(config)# *int e0* **Router(config-if)**# ip access-group 101 in

Extended IP Access List Port Operators

In the preceding example, we identified TCP port 80 on a specific host use the following syntax:

Router(config)# access-list 101 permit tcp 172.18.0.0 0.0.255.255 host 172.16.10.10 eq 80

We accomplished this using an operator of eq, which is short for equals. Thus, we are identifying host 172.16.10.10 with a port that equals 80.

We can use several other operators for port numbers:

eq	Matches a specific port
gt	Matches all ports greater than the port specified
lt	Matches all ports less than the port specified
neq	Matches all ports except for the port specified
range	Match a specific inclusive range of ports

The following will match all ports *greater* than 100:

Router(config)# access-list 101 permit tcp any host 172.16.10.10 gt 100

The following will match all ports *less* than 1024:

Router(config)# access-list 101 permit tcp any host 172.16.10.10 lt 1024

The following will match all ports that do *not equal 443*:

Router(config)# access-list 101 permit tcp any host 172.16.10.10 neg 443

The following will match all ports between 80 and 88:

Router(config)# access-list 101 permit tcp any host 172.16.10.10 range 80 88

Access List Logging

Consider again the following example:



Assume there is a webserver on the 172.16.x.x network with an IP address of 172.16.10.10.

We wish to keep track of the number of packets permitted or denied by each line of an access-list. Access-lists have a built-in logging mechanism for such a purpose:

Router(config)# access-list 101 permit tcp 172.18.0.0 0.0.255.255 host 172.16.10.10 eq 80 log Router(config)# access-list 101 deny ip 172.18.0.0 0.0.255.255 172.16.0.0 0.0.255.255 log **Router(config)**# access-list 101 permit ip any any log

Notice we added an additional keyword *log* to each line of the access-list. When viewing an access-list using the following command:

Router# show access-list 101

We will now have a counter on each line of the access-list, indicating the number of packets that were permitted or denied by that line. This information can be sent to a syslog server:

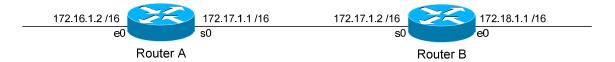
> Router(config)# logging on Router(config)# logging 172.18.1.50

The *logging on* command enables logging. The second *logging* command points to a syslog host at 172.18.1.50.

We can include more detailed logging information, including the source MAC address of the packet, and what interface that packet was received on. To accomplish this, use the *log-input* argument:

Router(config)# access-list 101 permit ip any any log-input

<u>ICMP Access List</u>



Consider this scenario. You've been asked to block anyone from the 172.18.x.x network from "pinging" anyone on the 172.16.x.x network. You want to allow everything else, including all other ICMP packets.

The specific ICMP port that a "ping" uses is **echo**. To block specific ICMP parameters, use an extended IP access list. On Router B, we would configure:

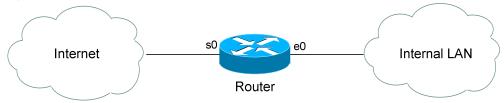
Router(config)# access-list 102 deny icmp 172.18.0.0 0.0.255.255 172.16.0.0 0.0.255.255 echo Router(config)# access-list 102 permit icmp 172.18.0.0 0.0.255.255 172.16.0.0 0.0.255.255 Router(config)# access-list 102 permit ip any any

The first line blocks only ICMP echo requests (pings). The second line allows all other ICMP traffic. The third line allows all other IP traffic.

Don't forget to apply it to an interface on Router B:

Router(config)# *int e0* **Router(config-if)**# ip access-group 102 in

Untrusted networks (such as the Internet) should usually be blocked from pinging an outside router or any internal hosts:



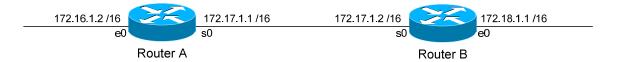
Router(config)# access-list 102 deny icmp any any **Router(config)**# access-list 102 permit ip any any

Router(config)# *interface s0* Router(config-if)# ip access-group 102 in

The above access-list completed disables ICMP on the serial interface. However, this would effectively disable ICMP traffic in both directions on the router. Any replies to pings initiated by the Internal LAN would be blocked on the way back in.

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Telnet Access List



We can create access lists to restrict telnet access to our router. For this example, we'll create an access list that prevents anyone from the evil 172.18.x.x network from telneting into Router A, but allow all other networks telnet access.

First, we create the access-list on Router A:

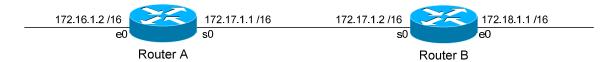
Router(config)# access-list 50 deny 172.18.0.0 0.0.255.255 **Router(config)**# access-list 50 permit any

The first line blocks the 172.18.x.x network. The second line allows all other networks.

To apply it to Router A's telnet ports:

Router(config)# line vty 0 4 Router(config-line)# access-class 50 in

Named Access Lists



Named access lists provide us with two advantages over numbered access lists. First, we can apply an identifiable name to an access list, for documentation purposes. Second, we can remove individual lines in a named access-list, which is not possible with numbered access lists.

Please note, though we can *remove* individual lines in a named access list, we cannot *insert* individual lines into that named access list. New entries are always placed at the bottom of a named access list.

To create a standard named access list, the syntax would be as follows:

Router(config)# ip access-list standard NAME Router(config-std-nacl)# deny 172.18.0.0 0.0.255.255 **Router(config-std-nacl)**# *permit any*

To create an extended named access list, the syntax would be as follows:

Router(config)# ip access-list extended NAME **Router(config-ext-nacl)**# permit tcp 172.18.0.0 0.0.255.255 host 172.16.10.10 eq 80 **Router(config-ext-nacl)**# deny ip 172.18.0.0 0.0.255.255 172.16.0.0 0.0.255.255 **Router(config-ext-nacl)**# *permit ip any any*

Notice that the actual configuration of the named access-list is performed in a separate router "mode":

Router(config-std-nacl)# Router(config-ext-nacl)#

Time-Based Access-Lists

Beginning with IOS version 12.0, access-lists can be based on the time and the day of the week.

The first step to creating a time-based access-list, is to create a *time-range*:

Router(config)# time-range BLOCKHTTP

The above command creates a *time-range* named *BLOCKHTTP*. Next, we must either specify an *absolute* time, or a *periodic* time:

Router(config)# time-range BLOCKHTTP
Router(config-time-range)# absolute start 08:00 23 May 2006 end 20:00 26 May 2006
Router(config)# time-range BLOCKHTTP
Router(config-time-range)# periodic weekdays 18:00 to 23:00

Notice the use of military time. The first *time-range* sets an *absolute* time that will *start* from May 23, 2006 at 8:00 a.m., and will *end* on May 26, 2006 at 8:00 p.m.

The second time-range sets a *periodic* time that is always in effect on *weekdays* from 6:00 p.m. to 11:00 p.m.

Only one *absolute* time statement is allowed per time-range, but multiple *periodic* time statements are allowed.

After we establish our time-range, we must reference it in an access-list:

Router(config)# access-list 102 deny any any eq 80 time-range BLOCKHTTP **Router(config)**# access-list 102 permit ip any any

Notice the *time-range* argument at the end of the access-list line. This will result in HTTP traffic being blocked, but only during the time specified in the time-range.

Source:

(http://www.cisco.com/univercd/cc/td/doc/product/software/ios120/120newft/120t/120t1/timerang.htm)

* * *

Advanced Wildcard Masks

Earlier in this section, we discussed the basics of wildcard masks. The examples given previously matched one of three things:

- A specific host
- A specific octet(s)
- All possible hosts

It is also possible to match groups or ranges of hosts with wildcard masks. For example, assume we wanted a standard access-list that denied the following hosts:

```
172.16.1.4
172.16.1.5
172.16.1.6
172.16.1.7
```

We could create an access-list with four separate lines:

```
Router(config)# access-list 10 deny 172.16.1.4 0.0.0.0
Router(config)# access-list 10 deny 172.16.1.5 0.0.0.0
Router(config)# access-list 10 deny 172.16.1.6 0.0.0.0
Router(config)# access-list 10 deny 172.16.1.7 0.0.0.0
```

However, it is also possible to match all four addresses in **one** line:

```
Router(config)# access-list 10 deny 172.16.1.4 0.0.0.3
```

How do I know this is correct? Let's write out the above four addresses, and my wildcard mask in binary:

172.16.1.4:	10101100.00010000.00000001.00000100
172.16.1.5:	10101100.00010000.00000001.00000101
172.16.1.6:	10101100.00010000.00000001.00000110
172.16.1.7:	10101100.00010000.00000001.00000111
Wild Card Mask:	0000000.00000000.00000000.00000011

Notice that the first 30 bits of each of the four addresses are identical. Each begin "10101100.00010000.00000001.000001". Since those bits must match exactly, the first 30 bits of our wildcard mask are set to **0**.

Advanced Wildcard Masks (continued)

Notice now that the *only* bits that are different between the four addresses are the last two bits. Not only that, but we use every computation of those last two bits: 00, 01, 10, 11.

Thus, since those last two bits can be anything, the last two bits of our wildcard mask are set to 1.

The resulting access-list line:

Router(config)# access-list 10 deny 172.16.1.4 0.0.0.3

We also could have determined the appropriate address and wildcard mask by using AND/XOR logic.

To determine the address, we perform a logical **AND** operation:

- 1. If all bits in a column are set to 0, the corresponding address bit is 0
- 2. If all bits in a column are set to 1, the corresponding address bit is 1
- 3. If the bits in a column are a mix of **0**'s and **1's**, the corresponding address bit is a **0**.

Observe:

172.16.1.4:	10101100.00010000.00000001.00000100
172.16.1.5:	10101100.00010000.00000001.00000101
172.16.1.6:	10101100.00010000.00000001.00000110
172.16.1.7:	10101100.00010000.00000001.00000111
Result:	10101100.00010000.00000001.00000100

Our resulting address is **172.16.1.4**. This gets us half of what we need.

Advanced Wildcard Masks (continued)

To determine the wildcard mask, we perform a logical **XOR** (exclusive OR) operation:

- 1. If all bits in a column are set to 0, the corresponding wildcard bit is 0
- 2. If all bits in a column are set to 1, the corresponding wildcard bit is 0
- 3. If the bits in a column are a mix of 0's and 1's, the corresponding wildcard bit is a 1.

Observe:

172.16.1.4:	10101100.00010000.00000001.00000100
172.16.1.5:	10101100.00010000.00000001.00000101
172.16.1.6:	10101100.00010000.00000001.00000110
172.16.1.7:	10101100.00010000.00000001.00000111
Result:	0000000.00000000.0000000.00000011

Our resulting wildcard mask is **0.0.0.3**. Put together, we have:

Router(config)# access-list 10 deny 172.16.1.4 0.0.0.3

Please Note: We can determine the number of addresses a wildcard mask will match by using a simple formula:

2ⁿ

Where "n" is the number of bits set to 1 in the wildcard mask. In the above example, we have two bits set to 1, which matches exactly **four addresses** $(2^2 = 4)$.

There will be occasions when we cannot match a range of addresses in one line. For example, if we wanted to deny 172.16.1.4-6, instead of 172.16.1.4-7, we would need two lines:

Router(config)# access-list 10 permit 172.16.1.7 0.0.0.0 **Router(config)**# access-list 10 deny 172.16.1.4 0.0.0.3

If we didn't include the first line, the second line would have denied the 172.16.1.7 address. Always remember to use the above formula (2ⁿ) to ensure your wildcard mask doesn't match more addresses than you intended (often called overlap).

Advanced Wildcard Masks (continued)

Two more examples. How would we deny all **odd** addresses on the 10.1.1.x/24 subnet in one access-list line?

Router(config)# access-list 10 deny 10.1.1.1 0.0.0.254

Written in binary:

10.1.1.1: 00001010.00000001.00000001.00000001 Wild Card Mask: 00000000.000000000.00000000.111111110

What would the result of the above wildcard mask be?

- 1. The first three octets must match exactly.
- 2. The last bit in the fourth octet must match exactly. Because we set this bit to 1 in our address, every number this matches will be **odd**.
- 3. All other bits in the fourth octet can match any number.

Simple, right? How would we deny all **even** addresses on the 10.1.1.x/24 subnet in one access-list line?

Router(config)# access-list 10 deny 10.1.1.0 0.0.0.254

Written in binary:

10.1.1.0: 00001010.00000001.00000001.00000000 Wild Card Mask: 00000000.000000000.00000000.111111110

What would the result of the above wildcard mask be?

- 4. The first three octets must match exactly.
- 5. The last bit in the fourth octet must match exactly. Because we set this bit to **0** in our address, every number this matches will be **even**.
- 6. All other bits in the fourth octet can match any number.

Section 23 - DNS and DHCP -

Name Resolution

Name resolution systems provide the translation between alphanumeric names and numerical addresses, alleviating the need for users and administrators to memorize long strings of numbers.

There are two common methods for implementing name resolution:

- A **static file** on each host on the network, containing all the name-toaddress translations (examples include the HOSTS/LMHOSTS files).
- A **centralized server** that all hosts on the network connect to for name resolution.

The two most common name resolution systems are **Domain Name System** (DNS) and Windows Internet Name Service (WINS). WINS was used in Microsoft networks to translate IP addresses to NetBIOS names, and is mostly deprecated.

DNS is heavily utilized on the Internet and on systems such as Active Directory.

Domain Name System (DNS)

Domain Name System (DNS) translates between *domain names* and *IP* addresses, and is supported by nearly every operating system. All Internetbased name resolution utilizes DNS.

DNS is organized as a hierarchy. Consider the following translation:

www.google.com = 209.85.225.104

The above domain name represents a **Fully Qualified Domain Name** (FQDN):

- .com represents a top level domain.
- .google represents a secondary level domain
- www represents a host computer in the .google.com domain.

Other top level domains include .org, .net, and .gov. Top level domains can also include country codes, such as .ca, .nl, and .de

Methods of configuring DNS

Recall that DNS name resolution can be implemented in the form of local HOSTS files, or a centralized name server(s). When employing **HOSTS** files, each translation must be statically configured on each device. In Windows 2000/XP operating systems, this file is located:

c:\windows\system32\drivers\etc\hosts

In UNIX/Linux operating systems, this file is generally located: /etc/hosts

There are many disadvantages to using HOSTS files. The HOSTS file must be configured on every device. If a change occurs, every device's HOSTS file must be updated.

Using one or more **DNS servers** provides several advantages over HOSTS files. All devices point to this centralized DNS server for name resolution, ensuring that changes only need to occur in one place.

If a particular DNS server does not contain the required DNS information, the request will can be *forwarded* to servers up the DNS hierarchy.

BIND (Berkeley Internet Name Domain) is the standard implementation of DNS. Microsoft, UNIX/Linux, and Novell all employ some version of BIND.

DNS servers assume one of three roles:

- Primary (or master) DNS Server maintains the SOA (Start of Authority), and contains the master zone file containing the DNS records for the domain. This server is often referred to as the Authoritative Name Server for a specific domain.
- Secondary (or slave) DNS Server maintains a current *copy* of the master zone file, obtained from the primary server. The secondary server cannot make changes to the zone file, but instead forwards changes to the primary server.
- Caching DNS Server does not maintain a zone file, and is not authoritative for any domain. This server will merely cache the results of DNS queries.

Both hosts and DNS servers will cache the result of DNS queries for a period of time.

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DNS Zone File Example

There are two types of **zones** in DNS:

- Forward Lookup Zones translates a host name to an IP address.
- **Reverse Lookup Zones -** translates an IP address to a hostname (otherwise known as the IN-ADDR.ARPA zone).

The following is an example zone file for the fictional *example.com* domain:

```
$ORIGIN example.com
$TTL 86400
      ΙN
              SOA
                      dns1.example.com.
hostmaster.example.com. (
                      2001062501 ; serial
                      21600 ; refresh after 6 hours
                      3600 ; retry after 1 hour
604800 ; expire after 1 week
86400 ) ; minimum TTL of 1 day
                      dns1.example.com.
       ΙN
              NS
              NS
                      dns2.example.com.
       ΙN
                      mail.example.com.
mail2.example.com.
       ΙN
              MΧ
              MX
       ΤN
              IN
IN
                     A
A
A
A
                               10.0.1.5
                               10.0.1.5
server1
server2
              IN
                               10.0.1.7
                             10.0.1.2
dns1
             ΤN
              IN
IN
IN
                      CNAME server1
mail
mail2
                      CNAME server2
WWW
                      CNAME server2
```

Entries within a zone file are referred to as DNS **records.** There are a variety of DNS record types, including:

- NS (Name Server) identifies a DNS server for the domain.
- **SOA** (**Start of Authority**) identifies the primary (authoritative) DNS server for the domain.
- A (Address) identifies an individual host in the domain.
- **CNAME** (Canonical Name) assigns an alias for another host name.
- MX (Mail Exchanger) identifies a mail server in the domain.
- **PTR** (**Pointer**) used for *reverse* DNS lookups.

The number defined in the MX record is a *priority*. A *lower* priority is more preferred.

DNS Process

DNS follows a strict process when performing a query. The process is as follows:

- 1. The *local DNS cache* on the host is queried first.
- 2. If there is no entry in the local cache, the *local HOSTS file* is aueried next.
- 3. If there is no entry in the local HOSTS, the query is forwarded to any configured DNS servers on the host. If no DNS servers are configured, the query will fail.
- 4. If the configured DNS server is not authoritative for that domain, and does not have that DNS entry locally cached, the query will be forwarded up the DNS hierarchy. DNS servers can be configured with one or more **forwarders**. Organizations often point to their ISP's DNS servers for DNS forwarding purposes.
- 5. If no forwarders are available, the query is forwarded to the *Root DNS server(s)*, which will likely have the entry cached.
- 6. In the rare circumstance that the Root servers do not have a cached entry, the query will be forwarded back down the hierarchy to the authoritative DNS server for that domain.

Dynamic DNS allows DNS to be integrated with Dynamic Host Configuration Protocol (DHCP). When DHCP hands out an IP address lease, it will automatically update the DNS entry for that host on the DNS server.

Resolving Hostnames on Cisco IOS Devices

There are two methods of name resolution on Cisco IOS devices:

- A static **host table** on each device (similar to a HOSTS file).
- A **centralized DNS server**(s) configured on each device.

To manually build a local host table on an IOS device:

Router(config)# ip host Router1 172.16.1.1 **Router(config)#** ip host Router2 172.17.1.2

To view the local host table:

Router# show hosts

To point an IOS device to a centralized DNS server:

Router(config)# *ip name-server 10.0.1.2*

To disable DNS lookups on an IOS device:

Router(config)# no ip domain-lookup

To configure the local domain on an IOS device:

Router(config)# ip domain-name CISCO.COM

DHCP (Dynamic Host Control Protocol)

In networks with a large number of hosts, statically assigning IP addresses and other IP information quickly becomes impractical.

Dynamic Host Control Protocol (DHCP) provides administrators with a mechanism to *dynamically* allocate IP addresses, rather than manually setting the address on each device.

DHCP servers **lease** out IP addresses to DHCP clients, for a specific period of time. There are four steps to this DHCP process:

- When a DHCP client first boots up, it broadcasts a **DHCPDiscover** message, searching for a DHCP server.
- If a DHCP server exists on the local segment, it will respond with a **DHCPOffer**, containing the "offered" IP address, subnet mask, etc.
- Once the client receives the offer, it will respond with a
 DHCPRequest, indicating that it will accept the offered protocol information.
- Finally, the server responds with a **DHCPACK**, acknowledging the clients acceptance of offered protocol information.

By default, DHCP leases an address for **8 days**. Once 50% of the lease expires, the client will try to renew the lease with the *same* DHCP server. If successful, the client receives a *new* 8 day lease.

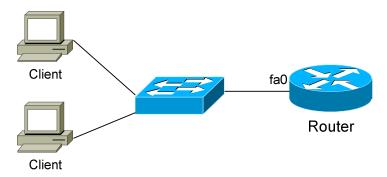
If the renewal is *not* successful, the client will continue "attempting" to renew, until 87.5% of the lease has expired. Once this threshold has been reached, the client will attempt to find another DHCP server to bind to.

In addition to IP address and subnet mask information, DHCP can provide the following protocol parameters:

- Default Gateway
- Domain Name and DNS servers
- Time Servers
- WINS servers

These are just a few examples of the many DHCP "options" that exist.

Configuring a Cisco Router as a DHCP Server



Cisco routers can be configured to function as DHCP servers. The first step is to create a DHCP **pool**:

> Router(config)# ip dhcp pool MYPOOL **Router(dhcp-config)#** network 192.168.1.0 255.255.255.0

The first command creates a *dhcp pool* named *MYPOOL*. The second command creates our DHCP scope, indicating the range of addresses to be leased. The above command indicates any address between 192.168.1.1 – 192.168.1.255 can be leased.

Specific addresses can be excluded from being leased:

Router(config)# ip dhcp excluded-address 192.168.1.1 Router(config)# ip dhcp excluded-address 192.168.1.5 192.168.1.10

The first command excludes only address 192.168.1.1. The second command excludes address 192.168.1.5 through 192.168.1.10.

To specify DHCP options to be leased with the address:

Router(config)# ip dhcp pool MYPOOL **Router(dhcp-config)**# default-router 192.168.1.1 Router(dhcp-config)# dns-server 192.168.1.5 Router(dhcp-config)# domain-name MYDOMAIN

To specify the duration of the DHCP lease:

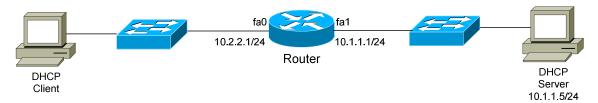
Router(config)# *ip dhcp pool MYPOOL* Router(dhcp-config)# lease 1 12

The above changes the default *lease* from 8 days to 1 day, 12 hours. To view current DHCP leases:

Router# show ip dhcp binding

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IP Helper Address



Recall that DHCP clients *broadcast* their **DHCPDiscover** packets, when searching for a DHCP server.

What would happen if the DHCP server is on a different network, separated from the clients by a router? Routers, by default, will never forward a broadcast.

Thus, in the above example, the client would never be able to reach the DHCP server to acquire its IP address. That is, unless the *ip helper-address* command is used:

Router(config)# interface fa0 **Router(config-if)#** ip helper-address 10.1.1.5

Notice that the *ip helper-address* command is configured on the interface connecting to the DHCP client, pointing to the IP address of the DHCP server. When the client broadcasts its DHCPDiscover packet, the router will direct that broadcast to the DHCP server. And there was much rejoicing.

By default, the *ip helper-address* command will forward the following UDP traffic:

- TFTP (port 69)
- DNS (port 53)
- Time (port 37)
- NetBIOS (ports 137-138)
- ND (Network Disks used by Sun workstations)
- TACACS (port 49)
- BOOTP/DHCP (ports 67-68)

Customized UDP traffic can be specified using the following command:

Router(config)# ip forward-protocol udp 107 **Router(config)**# no ip forward-protocol udp 69

(Reference: http://www.cisco.com/en/US/products/sw/iosswrel/ps5207/products_command_reference_chapter09186a0080238b72.html#wp1182972)

* * *

Part V

WANs

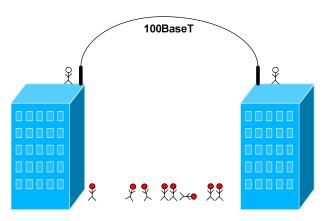
Section 24 - Basic WAN Concepts -

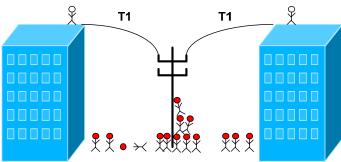
What is a WAN?

There are two prevailing definitions of a Wide Area Network (WAN). The **book definition** of a WAN is a network that spans large geographical locations, usually to interconnect multiple Local Area Networks (LANs).

The **practical definition** of a WAN is a network that traverses a public network or commercial carrier, using one of several WAN technologies.

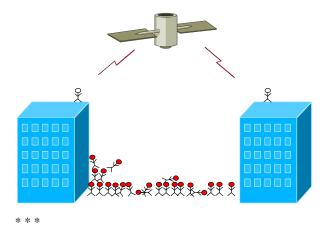
Consider the following example. A connection between two buildings using Ethernet as a medium would generally be considered a LAN. However, this is because of the technology used, and not the zombie-infested distance between the two buildings.





A connection between the same two buildings, using a dedicated T1 line as a medium, would generally be considered a WAN.

Remember, the difference is the technology used. A variety of WAN technologies exist, each operating at both the **Physical** and **Data-link** layers of the OSI models. Higher-layer protocols such as IP are **encapsulated** when sent across the WAN link.



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WAN Connection Types

WANs are generally grouped into three separate **connection types**:

- Point-to-Point technologies
- Circuit-switched technologies
- Packet-switched technologies

Point-to-Point technologies (often called **dedicated** or **leased lines**) are usually the most expensive form of WAN technology. Point-to-Point technologies are *leased* from a service provider, and provide *guaranteed* bandwidth from location to another (hence point-to-point). Cost is determined by the distance of the connection, and the amount of bandwidth allocated.

Generally, point-to-point links require **no call-setup**, and the connection is usually **always on**. Examples of point-to-point technologies include:

- T1 lines
- T3 lines

Circuit-Switched technologies require call-setup to occur before information can be transferred. The session is usually torn down once data transfer is complete (this is identified as an **On-Demand Circuit**). Circuitswitched lines are generally low-speed compared to point-to-point lines.

Examples of circuit-switched technologies include:

- Dial-up
- ISDN

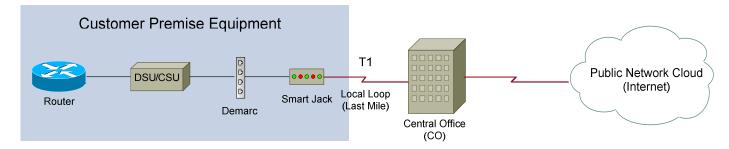
Packet-Switched technologies share a common infrastructure between all the provider's subscribers. Thus, bandwidth is not guaranteed, but is instead allocated on a **best effort** basis. Packet-switched technologies are ill-suited for applications that require consistent bandwidth, but are considerably less expensive than dedicated point-to-point lines.

Examples of packet-switched technologies include:

- Frame-Relay
- X25

(Reference: http://www.cisco.com/univercd/cc/td/doc/cisintwk/ito_doc/introwan.htm http://www.ciscopress.com/content/images/chap01 1587051486/elementLinks/1587051486content.pdf)

A wide variety of hardware is used with WANs. Equipment that is housed at the subscriber is referred to as **Customer Premise Equipment (CPE).**



The above example demonstrates the basic equipment required for a T1 line.

A CSU/DSU (Channelized Service Unit/Data Service Unit) provides the clocking and channelization for T1 or T3 technology. The CSU/DSU converts the signal for use on an Ethernet (or other LAN technology) network. If a WAN technology other than a T1 line is used, a different device will be required. Examples include (but are no limited to):

- ISDN a terminal adapter
- Dialup a modem

The **Demarc** (short for demarcation) refers to the point of last responsibility for the service provider. All equipment on the Customer Premises side of the Demarc is the customer's responsibility to maintain. The Demarc is not *always* physically labeled or identifiable. Occasionally, a two-port or fourport patch-panel will be used as a physical Demarc.

The **Smart Jack** physically terminates the T1 line. If there is a connectivity issue, the provider will perform a ping test to the smart jack. If communication to the smart jack is successful, the provider will assume the issue resides on the customer's side of responsibility. The smart jack is often locked in a glass enclosure, and labeled with the T1's circuit number.

The **Local Loop** (or Last Mile) refers to the physical line connecting from the Customer Premises to the provider's nearest **Central Office** (**CO**).

* * *

WAN Encapsulation

Recall that WAN technologies operate at both **Physical** and **Data-link** layers of the OSI models, and that higher-layer protocols such as IP are **encapsulated** when sent across the WAN link.

A WAN is usually terminated on a Cisco device's serial interface. Serial interfaces support a wide variety of WAN encapsulation types, which must be manually specified.

By default, a serial interface will utilize **HDLC** for encapsulation. Other supported encapsulation types include:

- SDLC
- PPP
- LAPB
- Frame-Relay
- X.25
- ATM

Regardless of the WAN encapsulation used, it must identical on both sides of a point-to-point link.

Each encapsulation type is described in detail in separate guides.

Section 25 - PPP -

WAN Encapsulation

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- LAPB
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- X.25
- ATM

Regardless of the WAN encapsulation used, it must **identical** on both sides of a point-to-point link.

HDLC Encapsulation

High-Level Data-link Control (HDLC) is a WAN encapsulation protocol used on dedicated point-to-point serial lines.

Though HDLC is technically an ISO standard protocol, Cisco's implementation of HDLC is proprietary, and will not work with other routers.

HDLC is also Cisco's **default encapsulation** type for serial point-to-point links. HDLC provides *no* authentication mechanism.

PPP Encapsulation

Point-to-Point Protocol (PPP) is a standardized WAN encapsulation protocol that can be used on a wide variety of WAN technologies, including:

- Serial dedicated point-to-point lines
- Asynchronous dial-up (essentially dialup)
- ISDN

PPP has four components:

- **EIA/TIA-232-C** standard for physical serial communication
- HDLC for encapsulating packets into frames over serial lines
- LCP for establishing, setting-up, and terminating point-to-point links
- NCP allows multiple Layer-3 protocols (such as IP and IPX) to be encapsulated into frames

PPP supports several features that HDLC does not:

- Authentication
- Compression
- Multi-link
- Error Control

Configuring PPP

To configure a serial interface for PPP encapsulation:

Router(config)# int s0/0
Router(config-if)# encapsulation ppp

PPP supports two methods of authentication, **PAP** and **CHAP**. PAP (**Password Authentication Protocol**) sends passwords in clear text, and thus does not provide much security. CHAP (**Challenge Handshake Authentication Protocol**) uses MD5 to apply an irreversible hash.

To configure PPP authentication:

Router(config)# hostname Router1
Router(config)# username Router2 password PASSWORD

Router(config)# *int s0/0* **Router(config-if)#** *ppp authentication chap*

The first line sets the hostname of the router. The second line sets the username and password used for PPP authentication. The username must be the hostname of the *remote* router, and the password must be the same on both routers.

The above configuration sets the authentication to *chap*. To instead configure *pap* authentication:

Router(config)# int s0/0
Router(config-if)# ppp authentication pap

To view the encapsulation configured on the interface:

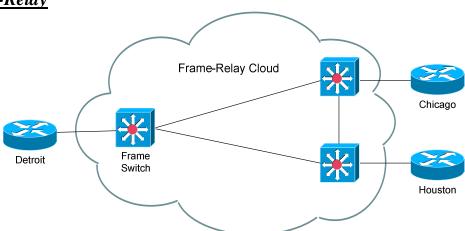
Router# show interface s0/0

To troubleshoot PPP authentication between two routers:

Router# debug ppp authentication

Section 26 - Frame-Relay -

Frame-Relay



Frame-Relay is a packet-switched technology, which shares bandwidth between users on the switched network. Frame-relay service providers assume that all clients will *not* need the full capacity of their bandwidth at all times. Thus, in general, frame-relay is less expensive than dedicated WAN lines, but customers are not guaranteed bandwidth.

All locations plug into the frame relay "cloud," which is a conglomeration of dozens or hundreds of Frame-Relay switches and routers. The cloud is the Frame provider's network, and the customer has no control (or even knowledge) of what occurs inside that infrastructure.

For communication to occur between locations, virtual circuits (VC) must be created. A VC is a one-way path through the Frame-Relay cloud.

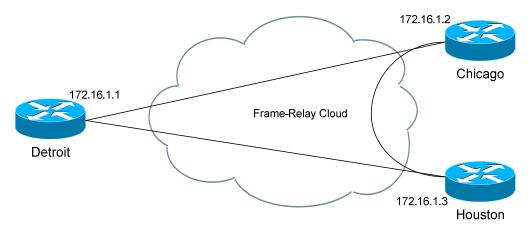
In the above example, in order to establish full communication between Detroit and Houston, we would need to create two virtual circuits:

- A virtual circuit between Detroit and Houston
- A separate virtual circuit between Houston and Detroit

Frame-relay circuits can either be **permanent** (**PVC**), or **switched** (**SVC**). A permanent virtual circuit is always kept active, and is the most common virtual circuit. A switched virtual circuit is created only when traffic needs to be sent, and is torn down when communication is complete.

Virtual circuits are identified with **Data Link Connection Identifiers (DLCIs).** Frame-Relay switches make decisions based on DLCIs, whereas Ethernet switches make decisions based on MAC addresses.

Frame-Relay Global vs Local DLCI



The difference between a globally or locally significant DLCI is all based on perspective. Remember that a DLCI identifies a one-way virtual circuit. For example, the connection between Detroit and Chicago would be considered one virtual circuit, and Chicago to Detroit would be a separate virtual circuit.

To get this to work, we need to map a DLCI to an IP address. For example, on router Detroit, we're going to create a virtual circuit to router Chicago. We'll assign it a DLCI of "102," and point it to Chicago's IP address.

We call this *locally* significant, because it only affects the interface on the Detroit router. We could, on the Chicago router, set a DLCI of "102" and point it to the IP address of the Detroit router. Because the DLCI is set on a different router (and interface), there will be no conflict.

When we set a *globally* significant DLCI, it is really only an administrative feature. It means that an administrator has consciously decided that all virtual circuits going to Chicago will be set to DLCI 102 (or whatever DLCI number you choose), whether it is from Detroit or Houston.

In essence, you are symbolically assigning the DLCI of 102 to the Chicago location. Keep in mind that you are still technically assigning the DLCI to the virtual circuits connecting to Chicago.

Virtual circuits pointing to other locations will be configured with *different* DLCIs (Detroit could be 101; Houston could be 103, etc.). The advantage to this is that it is now easy to determine the destination of a packet, based on its DLCI.

<u>Frame-Relay CIR</u>

Bandwidth is provided on a best effort basis in Frame-Relay.

The Frame provider and customer agree on a **Committed Information Rate** (CIR), which is not always a guarantee of bandwidth. The provider will give a best effort to meet the CIR, which is measured in bits per second:

- 256000 bps
- 512000 bps
- 1544000 bps

The above are examples of possible CIR settings, though technically the CIR can be set to anything. At times, bandwidth speeds can burst (Be) above the CIR. However, speeds above the CIR are certainly not guaranteed, and if the Frame Network becomes congested, any data exceeding the CIR becomes **Discard Eligible,** and is at risk of being dropped.

Frame-Relay Encapsulation Types

On Cisco routers, two possible Frame encapsulations can be configured on the router's serial ports.

- Cisco the default, and proprietary, Frame-Relay encapsulation
- **IETF** the standardized Frame-Relay encapsulation.

Frame-Relay Local Management Interface (LMI)

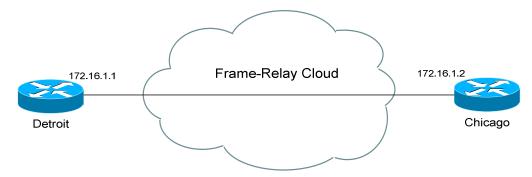
LMI is the type of signaling used between your router and your provider's Frame-Relay switch. LMI provides status updates of Virtual Circuits between the Frame switch and the router.

There are three LMI-types:

- **Cisco** default and proprietary (naturally)
- ANSI
- Q.933a

LMI type is auto-sensed on Cisco routers, but can be manually set if desired.

Frame-Relay Point-to-Point Configuration Example



Point-to-Point is the simplest form of Frame-Relay configuration. Remember that PVCs are only one-way circuits, and thus we need to create *two* PVCs in order for full communication to occur.

Configuration on the Detroit and Chicago routers would be as follows:

Detroit Router:

Router(config)# int s0/0 Router(config-if)# ip address 172.16.1.1 255.255.0.0 Router(config-if)# encapsulation frame-relay Router(config-if)# frame-relay lmi-type q933a Router(config-if)# frame-relay interface-dlci 102 Router(config-if)# no shut

Chicago Router:

Router(config)# int s0/0
Router(config-if) # ip address 172.16.1.2 255.255.0.0
Router(config-if)# encapsulation frame-relay
Router(config-if)# frame-relay lmi-type q933a
Router(config-if)# frame-relay interface-dlci 201
Router(config-if)# no shut

Notice that both routers are in the same IP subnet.

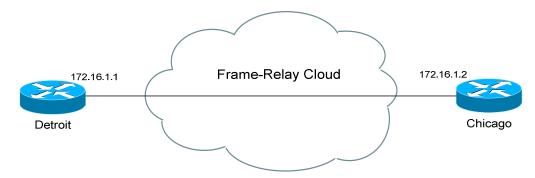
The *encapsulation frame-relay* command sets the frame encapsulation type to the default of *cisco*. The encapsulation must be the **same** on both routers. To change the default encapsulation type, simply append the *ietf* keyword to the *encapsulation frame-relay* command:

Router(config)# int s0/0
Router(config-if)# ip address 172.16.1.1 255.255.0.0
Router(config-if)# encapsulation frame-relay ietf

The *frame-relay lmi-type* command sets the signaling type. The Frame-Relay provider dictates which LMI-type to use. Remember that *cisco* is the default LMI-type, and that LMI is usually auto-sensed.

* * *

Frame-Relay Point-to-Point Configuration Example (continued)



Detroit Router:

Router(config)# int s0/0 **Router(config-if)**# ip address 172.16.1.1 255.255.0.0 **Router(config-if)**# encapsulation frame-relay Router(config-if)# frame-relay lmi-type q933a **Router(config-if)**# frame-relay interface-dlci 102 Router(config-if)# no shut

Chicago Router:

Router(config)# *int s0/0* **Router(config-if)**# ip address 172.16.1.2 255.255.0.0 **Router(config-if)**# encapsulation frame-relay Router(config-if)# frame-relay lmi-type q933a Router(config-if)# frame-relay interface-dlci 201

Router(config-if)# no shut

The frame-relay interface-dlci command identifies the one-way PVC. The connection between Detroit and Chicago has been assigned DLCI 102. The connection between Chicago and Detroit has been assigned DLCI 201.

The Frame-Relay provider usually dictates which DLCI numbers to use, as the provider's Frame switch is configured with the appropriate DLCI information.

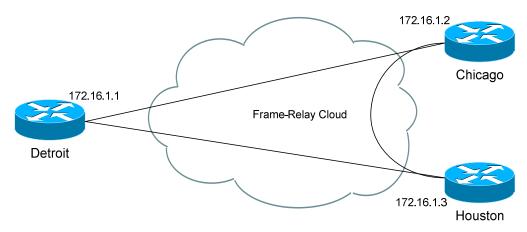
The router can actually receive all PVC and DLCI information directly from the Frame-Relay switch via LMI, using **Inverse-ARP**. Inverse-ARP is **enabled by default** on Cisco routers.

Thus, if the Frame-Relay switch is configured correctly, the *frame-relay* interface-dlci command could theoretically be removed, and the frame-relay connection will still work.

There are circumstances when DLCIs should be manually assigned. Inverse-ARP can be disabled on an interface with the following command:

> **Router(config)#** int s0/0 **Router(config-if)**# no frame-relay inverse-arp

Frame-Relay Full Mesh Configuration Example



Consider the above example, a full mesh between three locations. All routers can still belong to the same IP subnet; however, DLCI's must now be mapped to IP addresses, as multiple PVCs are necessary on each interface.

This can be dynamically configured via **Inverse-Arp**, which is enabled by default (as stated earlier). Otherwise, the DLCI-to-IP mapping can be performed manually. Looking at the Detroit and Chicago router's configuration:

Detroit Router:

Router(config-if)# no shut

Router(config)# int s0/0 **Router(config-if)**# ip address 172.16.1.1 255.255.0.0 Router(config-if)# encapsulation frame-relay ietf Router(config-if)# no frame-relay inverse-arp Router(config-if)# frame-relay lmi-type ansi Router(config-if)# frame-relay map ip 172.16.1.2 102 broadcast **Router(config-if)**# frame-relay map ip 172.16.1.3 103 broadcast

Chicago Router:

Router(config)# int s0/0
Router(config-if) # ip address 172.16.1.2 255.255.0.0
Router(config-if)# encapsulation frame-relay ietf
Router(config-if)# no frame-relay inverse-arp
Router(config-if)# frame-relay lmi-type ansi
Router(config-if)# frame-relay map ip 172.16.1.1 201 broadca
Router(config-if)# frame-relay map ip 172.16.1.3 203 broadca
Router(config-if)# no shut

Inverse-ARP was disabled using the *no frame-relay inverse-arp* command.

The frame-relay map command maps the remote router's IP address to a DLCI. On the Detroit router, a map was created to Chicago's IP (172.16.1.2), and that PVC was assigned a DLCI of 102. The broadcast option allows broadcasts and multicasts to be forwarded to that address, so that routing protocols such as OSPF can form neighbor relationships.

172.17.1.3

Houston

172.16.1.2 Chicago 172.16.1.1 Frame-Relay Cloud 172.17.1.1

Frame-Relay Partial Mesh Configuration Example

Full-mesh Frame-Relay environments can get quite expensive. Partial-mesh environments are often more cost-effective. A partial-mesh is essentially a hub-and-spoke design, with one central or hub location that all other locations must connect through.

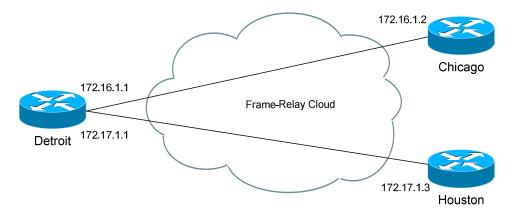
In the above example, the Detroit router serves as the hub router. In a partial-mesh environment, each **spoke** must be on a different IP subnet, which presents a special problem.

If both spokes terminate on the Detroit router's physical serial interface, split-horizon will prevent Chicago's routing updates from ever reaching Houston, and vice versa. Recall that split-horizon dictates that updates received on an interface cannot be sent back out the same interface.

Thus, on router Detroit, **sub-interfaces** must be created off of the serial interface. Sub-interfaces are virtual interfaces that the router treats as separate physical interfaces, providing a workaround for the split-horizon problem.

The **network type** must be specified when creating a sub-interface. A **pointto-point** sub-interface has only a single Virtual Circuit to another router. A multipoint sub-interface can have multiple Virtual Circuits to multiple locations.

Frame-Relay Partial Mesh Configuration Example (continued)



Configuration of the Detroit and Chicago routers would be as follows:

Detroit Router:

Router(config)# int s0/0 Router(config-if)# encapsulation frame-relay Router(config-if)# frame-relay lmi-type ansi

Router(config)# int s0/0.102 point-to-point **Router(config-subif)**# no frame-relay inverse-arp **Router(config-subif)**# *ip address* 172.16.1.1 255.255.0.0 Router(config-subif)# frame-relay interface-dlci 102 Router(config-subif)# no shut

Router(config)# int s0/0.103 point-to-point **Router(config-subif)**# no frame-relay inverse-arp **Router(config-subif)**# *ip address* 172.17.1.1 255.255.0.0 Router(config-subif)# frame-relay interface-dlci 103

Router(config-subif)# no shut

Chicago Router:

Router(config)# int s0/0 **Router(config-if)**# encapsulation frame-relay Router(config-if)# frame-relay lmi-type ansi

Router(config)# int s0/0.201 point-to-point Router(config-subif)# no frame-relay inverse-arp **Router(config-subif)**# ip address 172.16.1.2 255.255.0.0 Router(config-subif)# frame-relay interface-dlci 201 Router(config-subif)# no shut

Notice first that the Detroit router, serving as the hub, has two sub-interfaces configured pointing to Chicago and Houston. The Chicago router only has one sub-interface pointing to Detroit.

On the Detroit router, the *int s0/0.102* command creates a sub-interface numbered 102 on the Serial0/0 interface. Using the DLCI number for the sub-interface number is an arbitrary choice, useful for documentation purposes. On the Detroit router, each sub-interface contains only one virtual circuit, thus the interface's network type was set to *point-to-point*.

Notice also that encapsulation and LMI-type information is set on the physical interface, but IP address and DLCI information is set on the subinterface.

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Frame-Relay Traffic Shaping (FRTS)

Frame-Relay's method of QoS is called traffic-shaping, which controls the amount of traffic sent out an interface, and dictates congestion control mechanisms.

Frame-Relay Traffic-Shaping (FRTS) is used for two purposes:

- Adhering to the Frame provider's traffic rates.
- Preventing an oversubscription of the line between hub and spoke routers.

Several terms must be understood before configuring traffic-shaping:

- Committed Information Rate (CIR) the "average" traffic rate provided on a best-effort basis. By default, the CIR on a serial interface configured for traffic shaping is **56000** bits per second.
- Available Rate (AR) the maximum traffic rate, dictated either by the speed of the physical interface (using the *clock rate* command), or the restrictions of the Frame Provider.
- Minimum CIR (MinCIR) the minimum traffic rate the router will "throttle" down to if congestion occurs on the Frame-Relay network (i.e., a BECN is received). This is usually the provider's guaranteed traffic rate. By default, the MinCIR is *half* that of the CIR.
- **Discard Eligible (DE)** a bit that is set for all traffic sent above the MinCIR. Essentially, traffic that is sent above the Frame Provider's guaranteed rate can or will be dropped when congestion occurs.
- Committed Burst (Bc) the amount of bits sent during a specific interval, measured as **Time Committed** (**Tc**). To is measured in milliseconds (default is 125ms, or 8 intervals a second), and determines the number of intervals per second. The CIR is derived from the Bc and Tc using the following formula: $CIR = Bc \times 1000/Tc$
- Excess Burst (Be) the amount of bits that can be sent exceeding the Bc (or CIR). Any bits sent at this rate will be marked as DE.

Configuring Frame-Relay (FRTS)

To configure FRTS, a **map-class** must be created:

Router(config)# map-class frame-relay MYCLASS
Router(config-map-class)# frame-relay cir 64000
Router(config-map-class)# frame-relay bc 8000
Router(config-map-class)# frame-relay be 0
Router(config-map-class)# frame-relay mincir 32000
Router(config-map-class)# frame-relay adaptive-shaping becn

A *map-class* was created for *frame-relay* called *MYCLASS*. The first three commands configure the *CIR*, *Bc*, and *Be* respectively.

The final commands must be used in conjunction with each other. The *adaptive-shaping* feature has been specified, indicating that the router will throttle back to the *mincir* if a *becn* is received. The router does not throttle down to the *mincir* immediately, but rather will lower the rate by 25% until either the congestion stops, or the *mincir* is reached.

A map-class applied to an interface affects *all* PVCs on that interface. Additionally, map classes can be applied to a specific PVC, providing more granular control of FRTS.

To apply a map class to an interface:

Router(config)# interface s0/0
Router(config-if)# encapsulation frame-relay
Router(config-if)# frame-relay traffic-shaping
Router(config-if)# frame-relay class MYCLASS

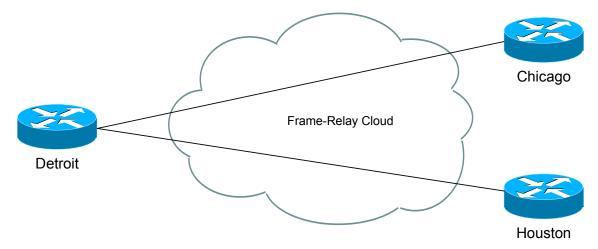
To apply a map class to a specific PVC:

Router(config)# interface s0/0
Router(config-if)# encapsulation frame-relay
Router(config-if)# frame-relay traffic-shaping
Router(config-if)# frame-relay interface-dlci 101 class MYCLASS

Do not forget the *frame-relay traffic-shaping* command. Once this command is configured, all PVCs are configured with the default CIR of 56,000 bps.

* * *

EIGRP and Frame-Relay



Observe the above Frame-Relay network. Two possible configuration options exist for the Detroit router:

- Configure frame-relay map statements on the physical interface
- Create separate sub-interfaces for each link, treating them as separate point-to-points.

If choosing the latter, EIGRP will treat each sub-interface as a separate link, and routing will occur with no issue.

If choosing the former, EIGRP will be faced with a split-horizon issue. Updates from Houston will not be forwarded to Chicago, and visa versa, as split horizon prevents an update from being sent out the link it was received on.

It is possible to *disable* split horizon for EIGRP:

Detroit(config)# interface s0/0 **Detroit(config-router)**# no ip split-horizon eigrp 10

Using sub-interfaces is Cisco's preferred method of circumventing the splithorizon issue, however.

Troubleshooting Frame-Relay

To view information concerning each PVC:

Router# *show frame-relay pvc*

The above command includes the following information:

- DLCI numbers
- Status of PVCs (active, inactive, deleted)
- Congestion information
- Traffic counters

To list Frame-Relay DLCI-mappings, whether manually created using the frame-relay map command, or created dynamically using Inverse ARP:

Router# show frame-relay map

To display the LMI-type configured on each interface, and LMI traffic statistics:

Router# show frame-relay lmi

To troubleshoot communication problems between the router and Frame-Relay switch:

Router# debug frame-relay lmi

To display information on packets *received* on a Frame-Relay interface:

Router# *debug frame-relay*

To display information on packets *sent* on a Frame-Relay interface:

Router# *debug frame-relay packet*

Section 27 - Network Address Translation -

NAT (Network Address Translation)

The rapid growth of the Internet resulted in a shortage of IPv4 addresses. In response, the powers that be designated a specific subset of the IPv4 address space to be *private*, to temporarily alleviate this problem.

A **public address** can be routed on the Internet. Thus, devices that should be Internet accessible (such as web or email servers) must be configured with public addresses.

A **private address** is only intended for use within an organization, and can never be routed on the internet. Three private addressing ranges were allocated, one for each IPv4 class:

- Class A 10.x.x.x
- Class B 172.16-31.x.x
- Class C 192.168.x.x

NAT (Network Address Translation) is used to translate between private addresses and public addresses. NAT allows devices configured with a private address to be *stamped* with a public address, thus allowing those devices to communicate across the Internet.

NAT is *not* restricted to just public-to-private address translations, though this is the most common application of NAT. NAT can perform a public-topublic address translation, or a private-to-private address translation as well.

NAT provides an additional benefit – hiding the specific addresses and addressing structure of the internal network.

Both Cisco IOS devices and PIX/ASA firewalls support NAT.

(Reference: http://www.cisco.com/en/US/tech/tk648/tk361/technologies_white_paper09186a0080194af8.shtml)

Types of NAT

NAT can be implemented using one of three methods:

Static NAT – performs a static one-to-one translation between two addresses, or between a *port* on one address to a port on another address. Static NAT is most often used to assign a public address to a device behind a NAT-enabled firewall/router.

Dynamic NAT – utilizes a **pool** of global addresses to dynamically translate the outbound traffic of clients behind a NAT-enabled device.

NAT Overload or Port Address Translation (PAT) – translates the outbound traffic of clients to unique port numbers off of a single global address. PAT is necessary when the number of internal clients exceeds the available global addresses.

NAT Terminology

Specific terms are used to identify the various NAT addresses:

- **Inside Local** the specific IP address assigned to an *inside* host behind a NAT-enabled device (usually a *private* address).
- **Inside Global** the address that identifies an *inside* host to the outside world (usually a public address). Essentially, this is the dynamically or statically-assigned public address assigned to a private host.
- Outside Global the address assigned to an *outside* host (usually a public address).
- **Outside Local** the address that identifies an *outside* host to the inside network. Often, this is the same address as the Outside Global. However, it is occasionally necessary to translate an outside (usually public) address to an inside (usually private) address.

For simplicity sake, it is generally acceptable to associate **global** addresses with **public** addresses, and **local** addresses with **private** addresses. However, remember that public-to-public and private-to-private translation is still possible. **Inside** hosts are within the local network, while **outside** hosts are external to the local network.



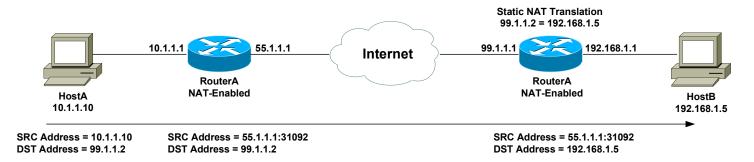
Consider the above example. For a connection *from* HostA *to* HostB, the NAT addresses are identified as follows:

- Inside Local Address 10.1.1.10
- Inside Global Address 55.1.1.1
- Outside Global Address 99.1.1.2
- Outside Local Address 99.1.1.2

HostA's configured address is 10.1.1.10, and is identified as its *Inside Local* address. When HostA communicates with the Internet, it is stamped with RouterA's public address, using PAT. Thus, HostA's *Inside Global* address will become 55.1.1.1.

When HostA communicates with HostB, it will access HostB's *Outside Global* address of 99.1.1.2. In this instance, the *Outside Local* address is also 99.1.1.2. HostA is never aware of HostB's configured address.

It is possible to map an address from the local network (such as 10.1.1.5) to the global address of the remote device (in this case, 99.1.1.2). This may be required if a legacy device exists that will only communicate with the local subnet. In this instance, the *Outside Local* address would be 10.1.1.5.



The above example demonstrates how the source (SRC) and destination (DST) IP addresses within the Network-Layer header are translated by NAT.

(Reference: http://www.cisco.com/warp/public/556/8.html)

* * *

Configuring Static NAT

The first step to configure **Static NAT** is to identify the *inside* (usually private) and *outside* (usually public) interfaces:

> **Router(config)#** *int e0/0* **Router(config)#** int s0/0 **Router**(config-if)# ip nat inside **Router(config-if)**# ip nat outside

To statically map a public address to a private address, the syntax is as follows:

Router(config)# ip nat inside source static 172.16.1.1 158.80.1.40

This command performs a *static* translation of the *source* address 172.16.1.1 (located on the *inside* of the network), to the outside address of 158.80.1.40.

Configuring Dynamic NAT

When configuring **Dynamic NAT**, the *inside* and *outside* interfaces must first be identified:

Router(config)# *int e0/0* **Router(config)#** int s0/0 **Router**(**config-if**)# *ip nat inside* **Router(config-if)**# *ip nat outside*

Next, a *pool* of global addresses must be specified. Inside hosts will dynamically choose the next available address in this pool, when communicating outside the local network:

Router(config)# ip nat pool POOLNAME 158.80.1.1 158.80.1.50 netmask 255.255.255.0

The above command specifies that the *pool* named *POOLNAME* contains a range of public addresses from 158.80.1.1 through 158.80.1.50.

Finally, a list of private addresses that are **allowed** to be dynamically translated must be specified:

> **Router(config)**# ip nat inside source list 10 pool POOLNAME **Router(config)**# access-list 10 permit 172.16.1.0 0.0.0.255

The first command states that any *inside* host with a *source* that matches access-list 10 can be translated to any address in the pool named POOLNAME.

The access-list specifies any host on the 172.16.1.0 network.

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Configuring NAT Overload (or PAT)

Recall that **NAT Overload** (or **PAT**) is necessary when the number of internal clients exceeds the available global addresses. Each internal host is translated to a unique port number off of a *single* global address.

Configuring NAT overload is relatively simple:

Router(config)# int e0/0
Router(config-if)# ip nat inside

Router(config)# int s0/0

Router(config-if)# ip nat outside

Router(config)# ip nat inside source list 10 interface Serial0/0 overload **Router(config)#** access-list 10 permit 172.16.1.0 0.0.0.255

Any *inside* host with a *source* that matches access-*list 10* will be translated with *overload* to the IP address configured on the *Serial0/0* interface.

Troubleshooting NAT

To view all current static and dynamic translations:

Router# show ip nat translations

To view whether an interface is configure as an *inside* or *outside* NAT interface, and to display statistical information regarding active NAT translations:

Router# *show ip nat statistics*

To view NAT translations in real-time:

Router# debug ip nat

To clear all dynamic NAT entries from the translation table:

Router# *clear ip nat translation*