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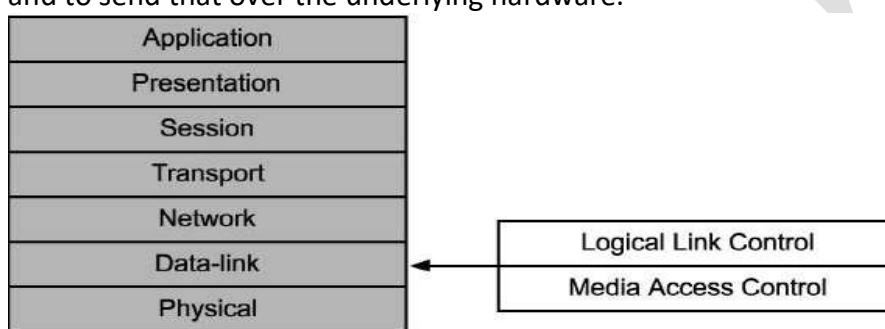
Department of Computer Science and Engineering  
CS602 Computer Networks  
Subject Notes: UNIT-II

**Syllabus:** Data Link Layer: Need, Services Provided, Framing, Flow Control, Error control. Data Link Layer Protocol: Elementary & Sliding Window protocol: 1-bit, Go-Back-N, Selective Repeat, Hybrid ARQ. Protocol verification: Finite State Machine Models & Petri net models. ARP/RARP/GARP

### DATA LINK LAYER: NEED

Data Link Layer is second layer of OSI Layered Model. This layer is one of the most complicated layers and has complex functionalities and liabilities. Data link layer hides the details of underlying hardware and represents itself to upper layer as the medium to communicate.

Data link layer works between two hosts which are directly connected in some sense. This direct connection could be point to point or broadcast. Data link layer is responsible for converting data stream to signals bit by bit and to send that over the underlying hardware.



**Fig. 2.1 Seven Layer Architecture**

Data link layer has two sub-layers:

- Logical Link Control: It deals with protocols, flow-control, and error control
- Media Access Control: It deals with actual control of media

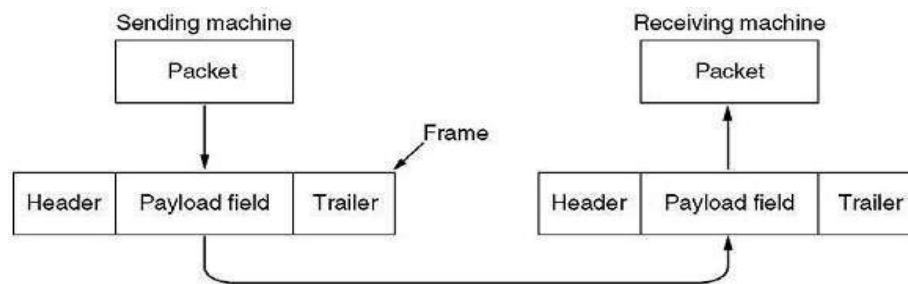
### DATA LINK LAYER: SERVICE PROVIDED

- Encapsulation of network layer data packets into frames.
- Frame synchronization.
- Error Control
- Flow control, in addition to the one provided on the transport layer.
- LAN switching (packet switching) including MAC filtering and spanning tree protocol
- Data packet queuing or scheduling
- Store-and-forward switching or cut-through switching

### DATA LINK LAYER: FRAMING

Since the physical layer merely accepts and transmits a stream of bits without any regard to meaning or structure, it is up to the data link layer to create and recognize frame boundaries. This can be accomplished by attaching special bit patterns to the beginning and end of the frame. If these bit patterns can accidentally occur in data, special care must be taken to make sure these patterns are not incorrectly interpreted as frame delimiters. The four framing methods that are widely used are

- Character count
- Starting and ending characters, with character stuffing
- Starting and ending flags, with bit stuffing



**Fig. 2.2 Data Link Layer: Framing**

### Character Count

This method uses a field in the header to specify the number of characters in the frame. When the data link layer at the destination sees the character count, it knows how many characters follow, and hence where the end of the frame is. The disadvantage is that if the count is garbled by a transmission error, the destination will lose synchronization and will be unable to locate the start of the next frame. So, this method is rarely used.

### Character stuffing

In the second method, each frame starts with the ASCII character sequence DLE STX and ends with the sequence DLE ETX. This method overcomes the drawbacks of the character count method. However, character stuffing is closely associated with 8-bit characters and this is a major hurdle in transmitting arbitrary sized characters.

### Bit stuffing

The third method allows data frames to contain an arbitrary number of bits and allows character codes with an arbitrary number of bits per character. At the start and end of each frame is a flag byte consisting of the special bit pattern 01111110. Whenever the sender's data link layer encounters five consecutive 1s in the data, it automatically stuffs a zero bit into the outgoing bit stream. This technique is called bit stuffing.

### Physical layer coding violations

The final framing method is physical layer coding violations and is applicable to networks in which the encoding on the physical medium contains some redundancy. In such cases normally, a 1 bit is a high-low pair and a 0 bit is a low-high pair. The combinations of low-low and high-high which are not used for data may be used for marking frame boundaries.

## DATALINK LAYER: FLOW CONTROL

Flow control coordinates that amount of data that can be sent before receiving acknowledgement.

- It is one of the most important duties of the data link layer.
- Flow control tells the sender how much data to send.
- It makes the sender wait for some sort of an acknowledgment (ACK) before continuing to send more data.
- Flow Control Techniques: Stop-and-wait, and Sliding Window

## DATA LINK LAYER: ERROR CONTROL

Error control in the data link layer is based on ARQ (automatic repeat request), which is the retransmission of data.

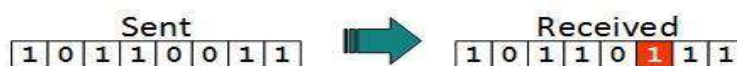
- The term error control refers to methods of error detection and retransmission.
- Anytime an error is detected in an exchange, specified frames are retransmitted. This process is called ARQ.

To ensure reliable communication, there needs to exist flow control (managing the amount of data the sender sends), and error control (that data arrives at the destination error free).

- Flow and error control needs to be done at several layers.
- For node-to-node links, flow and error control is carried out in the data-link layer.

- For end-point to end-point, flow and error control is carried out in the transport layer.

There may be three types of errors:



**Fig. 2.3 Single bit error**

In a frame, there is only one bit, anywhere though, which is corrupt.



**Fig. 2.4 Multiple bits error**

Frame is received with more than one bit in corrupted state.



**Fig. 2.5 Burst error**

Frame contains more than 1 consecutive bits corrupted.

## DATA LINK LAYER PROTOCOL

The basic function of the layer is to transmit frames over a physical communication link. Transmission may be half duplex or full duplex. To ensure that frames are delivered free of errors to the destination station (IMP) a number of requirements are placed on a data link protocol. The protocol (control mechanism) should be capable of performing:

1. The identification of a frame (i.e. recognises the first and last bits of a frame).
2. The transmission of frames of any length up to a given maximum. Any bit pattern is permitted in a frame.
3. The detection of transmission errors.
4. The retransmission of frames which were damaged by errors.
5. The assurance that no frames were lost.
6. In a multidrop configuration some mechanism must be used for preventing conflicts caused by simultaneous transmission by many stations.
7. The detection of failure or abnormal situations for control and monitoring purposes.

It should be noted that as far as layer 2 is concerned a host message is pure data, every single bit of which is to be delivered to the other host. The frame header pertains to layer 2 and is never given to the host.

## Elementary Data Link Protocols

- Data are transmitted in one direction only
- The transmitting (Tx) and receiving (Rx) hosts are always ready
- Processing time can be ignored
- Infinite buffer space is available
- No errors occur; i.e. no damaged frames and no lost frames (perfect channel)

## Sliding Window protocol:

A sliding window protocol is a feature of packet-based data transmission protocols. Sliding window protocols are used where reliable in-order delivery of packets is required, such as in the Data Link Layer (OSI model) as well as in the Transmission Control Protocol (TCP).

The Sliding Window ARQ has three techniques

1. 1-bit
2. Go- Back N
3. Selective Repeat

### 1-bit

One-bit sliding window protocol is also called Stop-And-Wait protocol. In this protocol, the sender sends out one frame, waits for acknowledgment before sending next frame, thus the name Stop-And-Wait.

Problem with Stop-And-Wait protocol is that it is very inefficient. At any one moment, only in frame is in transition. The sender will have to wait at least one round trip time before sending next. The waiting can be long for a slow network such as satellite link.

### Stop and Wait Protocol

#### Characteristics

- Used in Connection-oriented communication.
- It offers error and flow control
- It is used in Data Link and Transport Layers
- Stop and Wait ARQ mainly implements Sliding Window Protocol concept with Window Size 1

#### Useful Terms:

- **Propagation Delay:** Amount of time taken by a packet to make a physical journey from one router to another router.

$$\text{Propagation Delay} = (\text{Distance between routers}) / (\text{Velocity of propagation})$$

- RoundTripTime (RTT) =  $2 * \text{Propagation Delay}$
- TimeOut (TO) =  $2 * \text{RTT}$
- Time To Live (TTL) =  $2 * \text{TimeOut}$ . (Maximum TTL is 180 seconds)

### Simple Stop and Wait

#### Sender:

- Rule 1) Send one data packet at a time.
- Rule 2) Send next packet only after receiving acknowledgement for previous.

#### Receiver:

- Rule 1) Send acknowledgement after receiving and consuming of data packet.
- Rule 2) after consuming packet acknowledgement need to be sent (Flow Control)

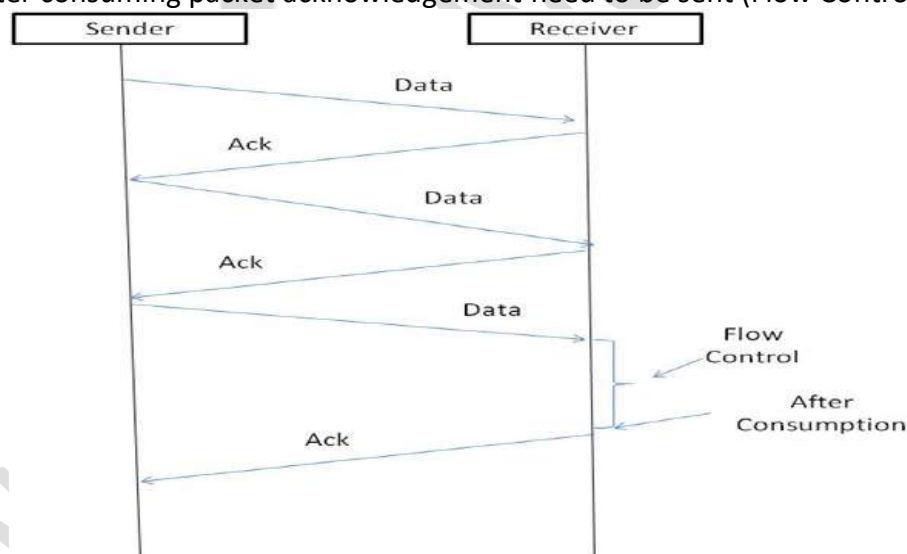


Fig. 2.6 Stop and Wait

#### Problems:

##### 1. Lost Data

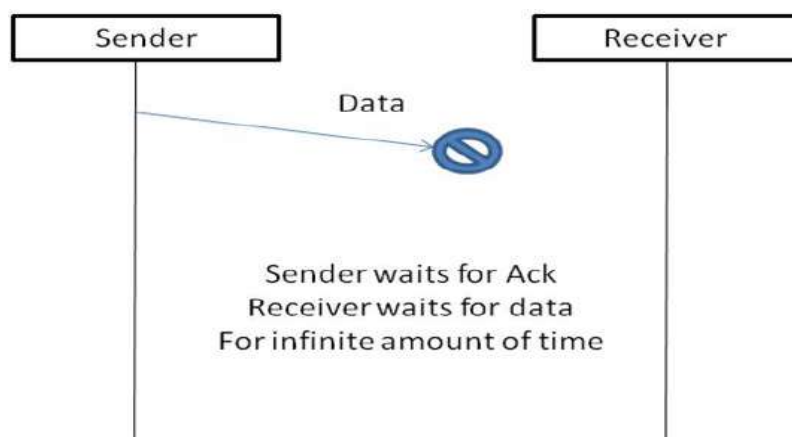


Fig. 2.7 Stop and Wait- Lost Data

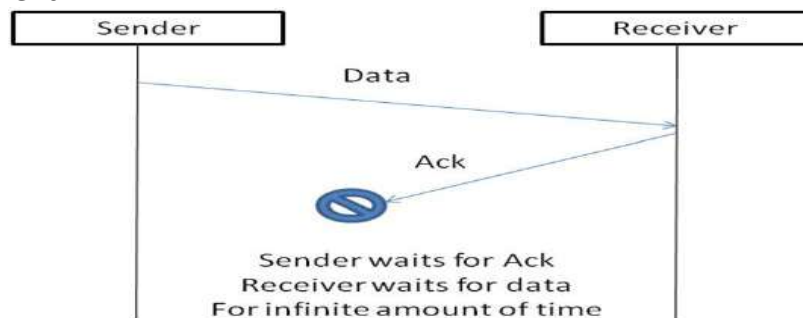
**2. Lost Acknowledgement:**

Fig. 2.8 Stop and Wait- Lost Acknowledgement

**3. Delayed Acknowledgement/Data:** After timeout on sender side, a long-delayed acknowledgement might be wrongly considered as acknowledgement of some other recent packet.

**Stop and Wait ARQ (Automatic Repeat Request)**

Above 3 problems are resolved by Stop and Wait ARQ (Automatic Repeat Request) that does both error control and flow control.

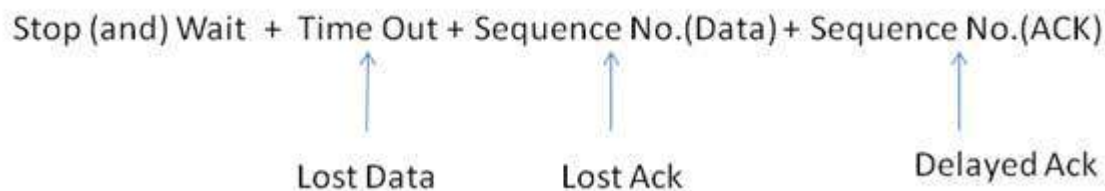


Fig. 2.9 Stop and Wait ARQ (Automatic Repeat Request)

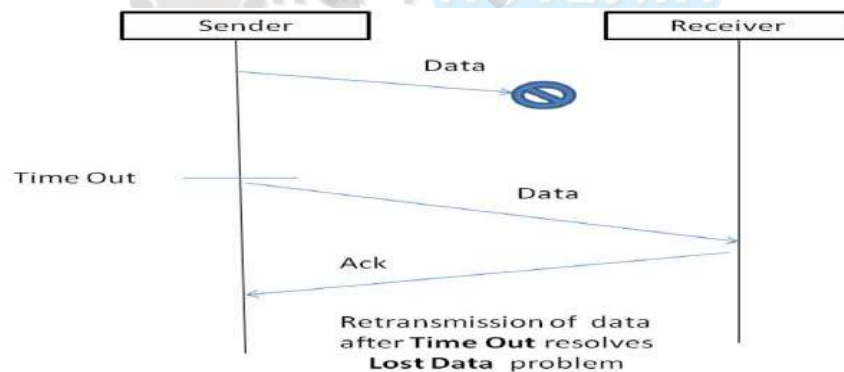
**1. Time Out:**

Fig. 2.10 Stop and Wait ARQ-Time Out

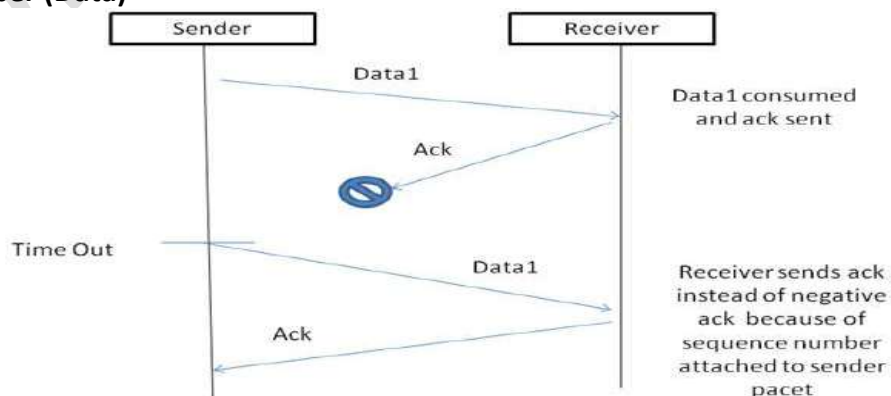
**2. Sequence Number (Data)**

Fig. 2.11 Stop and Wait ARQ-ACK Lost

**3. Delayed Acknowledgement:**

This is resolved by introducing sequence number for acknowledgement also.



### Working of Stop and Wait ARQ:

1) Sender A sends a data frame or packet with sequence number 0.

2) Receiver B, after receiving data frame, sends an acknowledgement with sequence number 1 (sequence number of next expected data frame or packet)

There is only one-bit sequence number that implies that both sender and receiver have buffer for one frame or packet only.

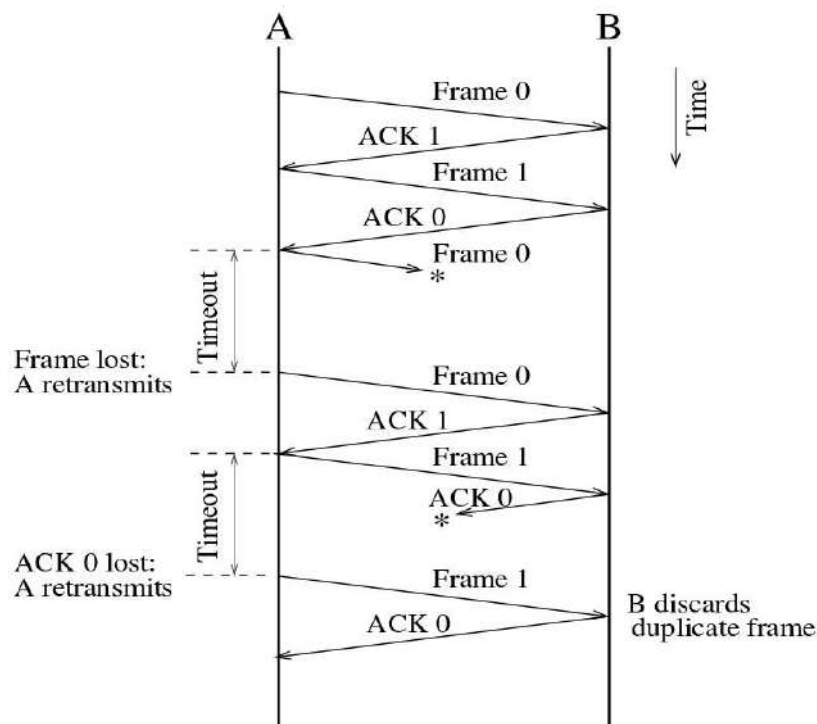


Fig. 2.12 Working of Stop and Wait ARQ

### Characteristics of Stop and Wait ARQ:

- It uses link between sender and receiver as half duplex link
- Throughput = 1 Data packet/frame per RTT
- If Bandwidth\*Delay product is very high, then stop and wait protocol is not so useful. The sender has to keep waiting for acknowledgements before sending the processed next packet.
- It is an example for “**Closed Loop OR connection oriented**” protocols
- It is a special category of SWP where its window size is 1
- Irrespective of number of packets sender is having stop and wait protocol requires only 2 sequences numbers 0 and 1

The Stop and Wait ARQ solves main three problems, but may cause big performance issues as sender always waits for acknowledgement even if it has next packet ready to send. Consider a situation where you have a high bandwidth connection and propagation delay is also high (you are connected to some server in some other country though a high-speed connection). To solve this problem, we can send more than one packet at a time with a larger sequence numbers. We will be discussing these protocols in next articles.

So, Stop and Wait ARQ may work fine where propagation delay is very less for example LAN connections, but performs badly for distant connections like satellite connection.

### Go-Back N protocol

Go-Back-N protocol is a sliding window protocol. It is a mechanism to detect and control the error in datalink layer. During transmission of frames between sender and receiver, if a frame is damaged, lost, or an acknowledgement is lost then the action performed by sender and receiver.

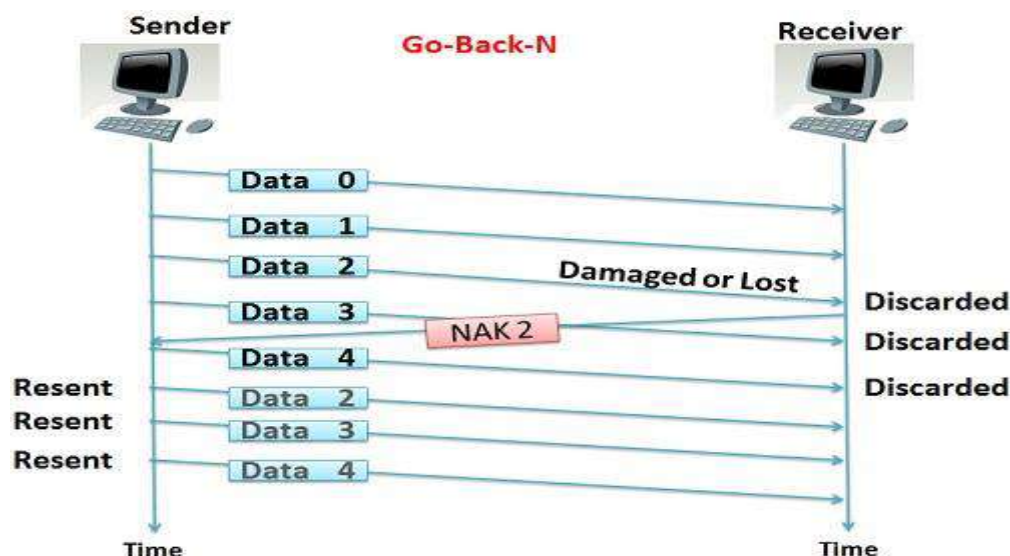


Fig. 2.13 Go- Back N protocol

### Selective Repeat protocol

Selective repeat is also the sliding window protocol which detects or corrects the error occurred in datalink layer. The selective repeat protocol retransmits only that frame which is damaged or lost. In selective repeat protocol, the retransmitted framed is received out of sequence. The selective repeat protocol can perform following actions

- The receiver is capable of sorting the frame in a proper sequence, as it receives the retransmitted frame whose sequence is out of order of the receiving frame.
- The sender must be capable of searching the frame for which the NAK has been received.
- The receiver must contain the buffer to store all the previously received frame on hold till the retransmitted frame is sorted and placed in a proper sequence.
- The ACK number, like NAK number, refers to the frame which is lost or damaged.
- It requires the less window size as compared to go-back-n protocol.

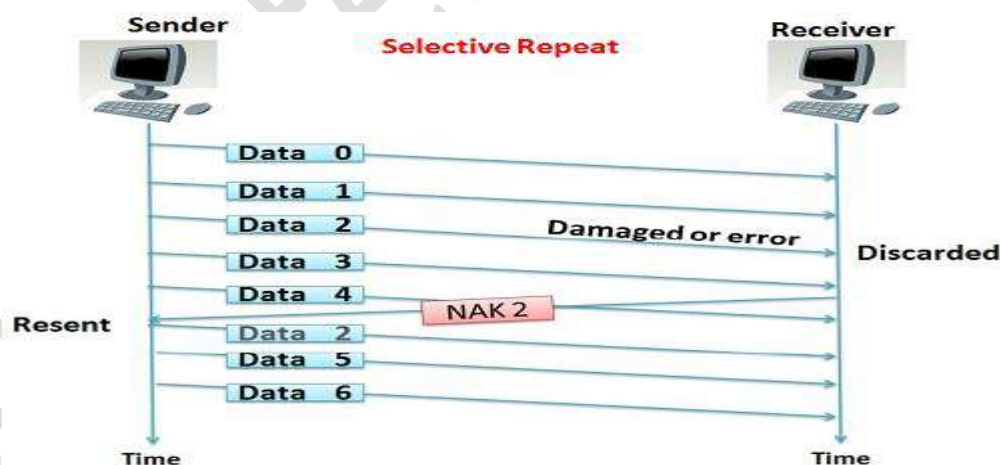


Fig. 2.14 Selective Repeat protocol

### HYBRID ARQ

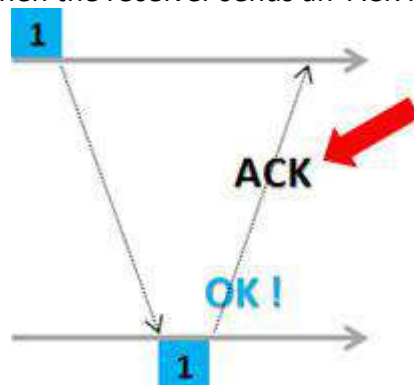
The HARQ is the use of conventional ARQ along with an Error Correction technique called 'Soft Combining', which no longer discards the received bad data (with error).

With the 'Soft Combining' data packets that are not properly decoded are not discarded anymore. The received signal is stored in a 'buffer', and will be combined with next retransmission.

That is, two or more packets received each one with insufficient SNR to allow individual decoding can be combined in such a way that the total signal can be decoded!

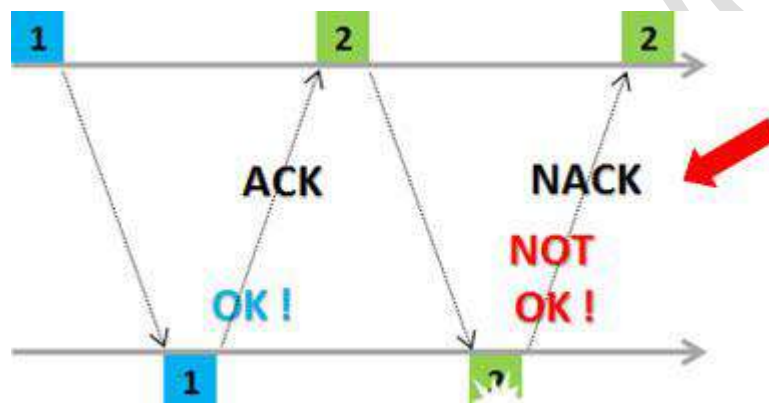


The following image explains this procedure. The transmitter sends a package [1]. The package [1] arrives, and is 'OK'. If the package [1] is 'OK' then the receiver sends an 'ACK'.



**Fig. 2.15 Transmitter sends a packet-1**

The transmission continues, and is sent a package [2]. The package [2] arrives, but let's consider now that it arrives with errors. If the package [2] arrives with errors, the receiver sends a 'NACK'.



**Fig. 2.16 Transmitter sends a packet-2**

Only now this package [2] (bad) is not thrown away, as it is done in conventional ARQ. Now it is stored in a 'buffer'.

buffer ?

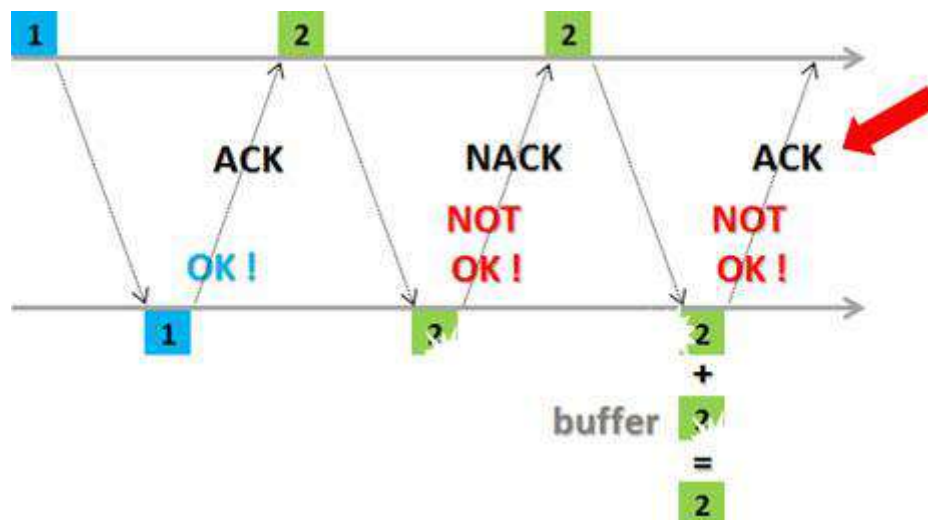
**Fig. 2.17 Receiver buffers a packet-2**

Continuing, the transmitter sends another package [2.1] that also (let's consider) arrives with errors.



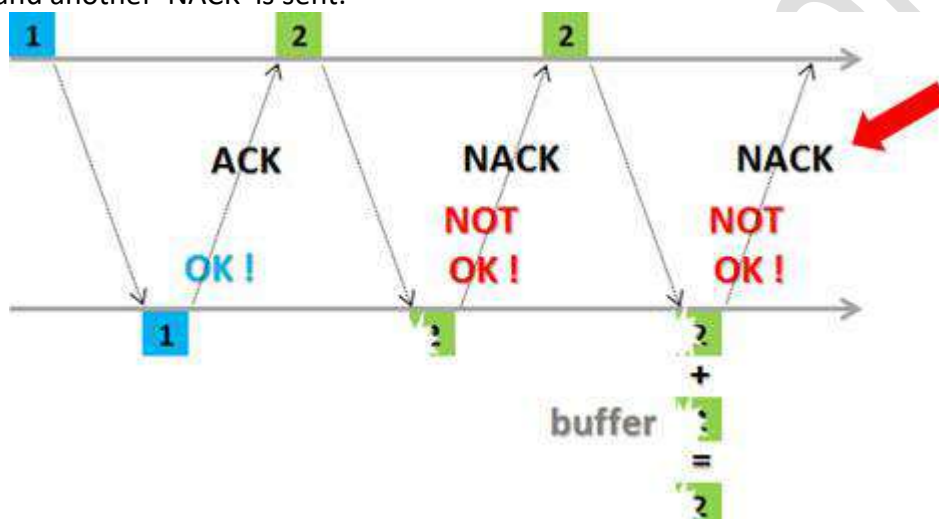
**Fig. 2.18 Transmitter sends another packet-2**

We have then in a buffer: bad package [2], and another package [2.1] which is also bad. Does by adding (combining) these two packages ([2] + [2.1]) we have the complete information? Yes. So, we send an 'ACK'.



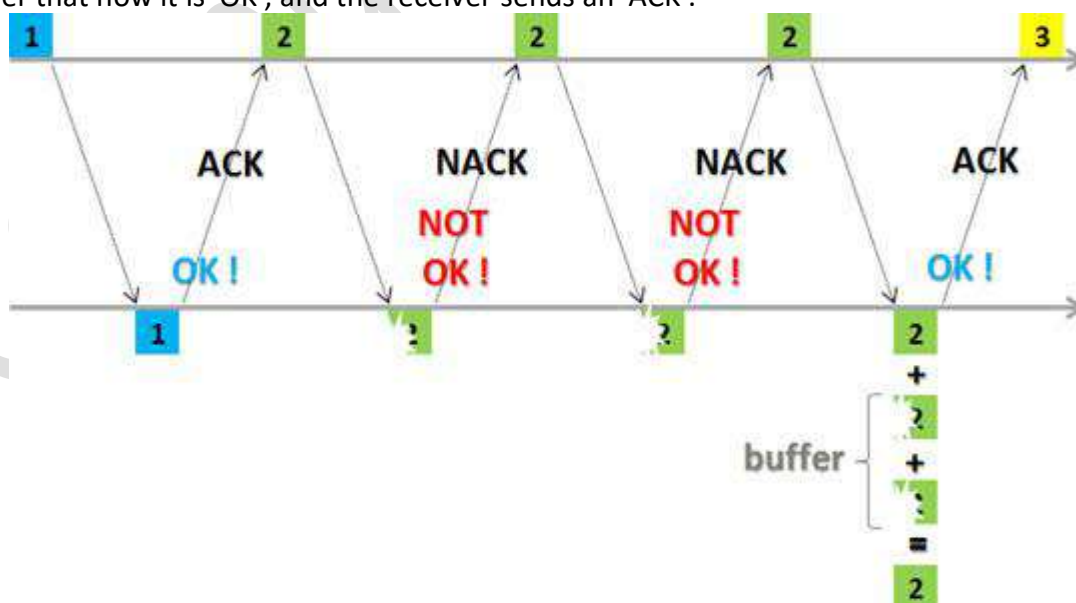
**Fig. 2.19 Receiver combining buffers a packet-2 and another packet-2**

But if the combination of these two packages still does not give us the complete information, the process must continue - and another 'NACK' is sent.



**Fig. 2.20 Receiver sends NACK**

And there we have another retransmission. Now the transmitter sends a third package [2.2]. Let's consider that now it is 'OK', and the receiver sends an 'ACK'.



**Fig. 2.21 Receiver sends ACK**

Here we can see the following: along with the received package [2.2], the receiver also has packages [2] and [2.1] that have not been dropped and are stored in the buffer.

In our example, we see that the package arrived 2 times 'wrong'. And what is the limit of these retransmissions? Up to 4. IE, we can have up to 4 retransmissions in each process. This is the maximum number supported by 'buffer'.

## BIT ORIENTED PROTOCOLS

A bit-oriented protocol is a communications protocol that sees the transmitted data as an opaque stream of bits with no semantics, or meaning. Control codes are defined in terms of bit sequences instead of characters. Bit oriented protocol can transfer data frames regardless of frame contents. It can also be stated as "bit stuffing" this technique allows the data frames to contain an arbitrary number of bits and allows character codes with arbitrary number of bits per character.

### SDLC

Synchronous Data Link Control (SDLC) supports a variety of link types and topologies. It can be used with point-to-point and multipoint links, bounded and unbounded media, half-duplex and full-duplex transmission facilities, and circuit-switched and packet-switched networks.

SDLC identifies two types of network nodes: primary and secondary. Primary nodes control the operation of other stations, called secondary. The primary polls the secondary in a predetermined order and secondary can then transmit if they have outgoing data. The primary also sets up and tears down links and manages the link while it is operational. Secondary nodes are controlled by a primary, which means that secondary can send information to the primary only if the primary grants permission.

### SDLC primaries and secondary can be connected in four basic configurations:

- Point-to-point---Involves only two nodes, one primary and one secondary.
- Multipoint---Involves one primary and multiple secondary.
- Loop---Involves a loop topology, with the primary connected to the first and last secondary. Intermediate secondary pass messages through one another as they respond to the requests of the primary.
- Hub go-ahead---Involves an inbound and an outbound channel. The primary uses the outbound channel to communicate with the secondary. The secondary use the inbound channel to communicate with the primary. The inbound channel is daisy-chained back to the primary through each secondary.

### SDLC Frame Format

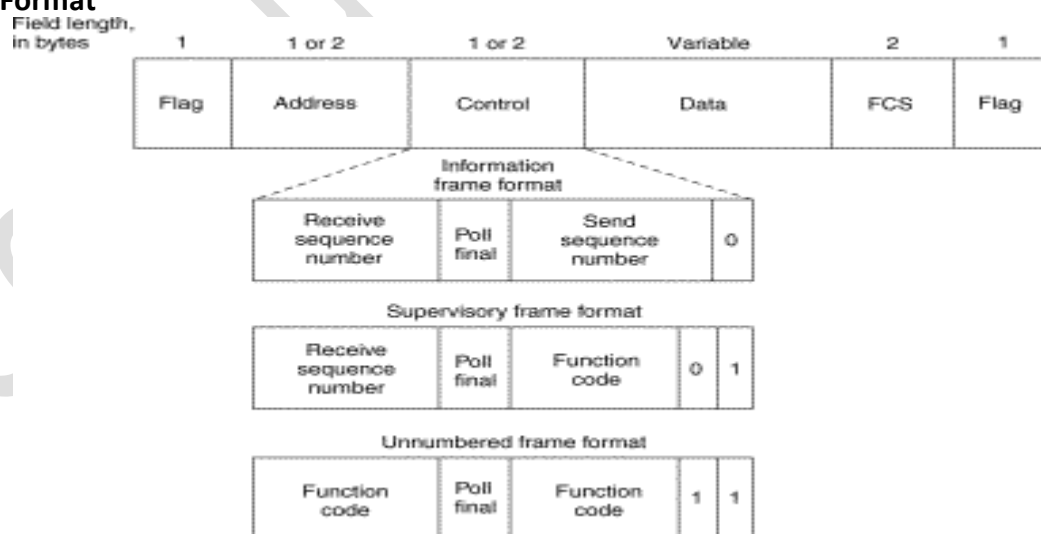


Fig. 2.22 SDLC Frame Format

- **Flag**---Initiates and terminates error checking.
- **Address**---Contains the SDLC address of the secondary station, which indicates whether the frame comes from the primary or secondary. This address can contain a specific address, a group address, or a broadcast address. A primary is either a communication source or a destination, which eliminates the need to include the address of the primary.
- **Control**---Employs three different formats, depending on the type of SDLC frame used:

1. **Information (I) frame:** Carries upper-layer information and some control information. This frame sends and receives sequence numbers, and the poll final (P/F) bit performs flow and error control. The send-sequence number refers to the number of the frame to be sent next. The receive-sequence number provides the number of the frame to be received next. Both sender and receiver maintain send- and receive-sequence numbers.  
A primary station uses the P/F bit to tell the secondary whether it requires an immediate response. A secondary station uses the P/F bit to tell the primary whether the current frame is the last in its current response.
  2. **Supervisory (S) frame:** Provides control information. An S frame can request and suspend transmission, reports on status, and acknowledge receipt of I frames. S frames do not have an information field.
  3. **Unnumbered (U) frame:** Supports control purposes and is not sequenced. A U frame can be used to initialize secondary. Depending on the function of the U frame, its control field is 1 or 2 bytes. Some U frames have an information field.
- **Data**---Contains path information unit (PIU) or exchange identification (XID) information.
  - **Frame Check Sequence (FCS)** ---Precedes the ending flag delimiter and is usually a cyclic redundancy check (CRC) calculation remainder. The CRC calculation is redone in the receiver. If the result differs from the value in the original frame, an error is assumed.

## HDLC

High-Level Data Link Control (HDLC) is a bit-oriented code-transparent synchronous data link layer protocol. HDLC provides both connection-oriented and connectionless service. HDLC can be used for point to multipoint connections, but is now used almost exclusively to connect one device to another, using what is known as Asynchronous Balanced Mode (ABM). The original master-slave modes Normal Response Mode (NRM) and Asynchronous Response Mode (ARM) are rarely used.

## FRAMING

HDLC frames can be transmitted over synchronous or asynchronous serial communication links. Those links have no mechanism to mark the beginning or end of a frame, so the beginning and end of each frame has to be identified. This is done by using a frame delimiter, or flag, which is a unique sequence of bits that is guaranteed not to be seen inside a frame. This sequence is '01111110', or, in hexadecimal notation, 0x7E. Each frame begins and ends with a frame delimiter. A frame delimiter at the end of a frame may also mark the start of the next frame. A sequence of 7 or more consecutive 1-bits within a frame will cause the frame to be aborted.

When no frames are being transmitted on a simplex or full-duplex synchronous link, a frame delimiter is continuously transmitted on the link. Using the standard NRZI encoding from bits to line levels (0 bit = transition, 1 bit = no transition), this generates one of two continuous waveforms, depending on the initial state:

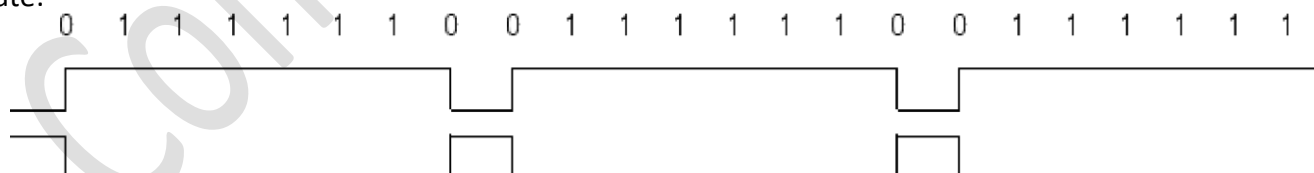


Fig. 2.23 HDLC Framing

This is used by modems to train and synchronize their clocks via phase-locked loops. Some protocols allow the 0-bit at the end of a frame delimiter to be shared with the start of the next frame delimiter, i.e. '011111101111110'.

## Frame structure

The contents of an HDLC frame are shown in the following table:

Flag	Address	Control	Information	FCS	Flag
8 bits	8 or more bits	8 or 16 bits	Variable length, 0 or more bits	16 or 32 bits	8 bits

**Fig. 2.24 HDLC Frame structure**

Note that the end flag of one frame may be (but does not have to be) the beginning (start) flag of the next frame.

Data is usually sent in multiples of 8 bits, but only some variants require this; others theoretically permit data alignments on other than 8-bit boundaries.

There are three fundamental types of HDLC frames.

- Information frames, or I-frames, transport user data from the network layer. In addition, they can also include flow and error control information piggybacked on data.
- Supervisory Frames, or S-frames, are used for flow and error control whenever piggybacking is impossible or inappropriate, such as when a station does not have data to send. S-frames do not have information fields.
- Unnumbered frames, or U-frames, are used for various miscellaneous purposes, including link management. Some U-frames contain an information field, depending on the type.

### BISYNC

Binary Synchronous Communication (BSC or Bisync) is an IBM character-oriented, half duplex link protocol, announced in 1967 after the introduction of System/360. It replaced the synchronous transmit-receive (STR) protocol used with second generation computers. The intent was that common link management rules could be used with three different character encodings for messages. Six-bit Transcode looked backwards to older systems.

BISYNC establishes rules for transmitting binary-coded data between a terminal and a host computer's BISYNC port. While BISYNC is a half-duplex protocol, it will synchronize in both directions on a full-duplex channel. BISYNC supports both point-to-point (over leased or dial-up lines) and multipoint transmissions. Each message must be acknowledged, adding to its overhead.

BISYNC is character oriented, meaning that groups of bits (bytes) are the main elements of transmission, rather than a stream of bits. The BISYNC frame is pictured next. It starts with two sync characters that the receiver and transmitter use for synchronizing. This is followed by a start of header (SOH) command, and then the header. Following this are the start of text (STX) command and the text. Finally, an end of text (EOT) command and a cyclic redundancy check (CRC) end the frame. The CRC provides error detection and correction.

**Fig. 2.25 BISYNC**

Most of the bisynchronous protocols, of which there are many, provide only half-duplex transmission and require an acknowledgment for every block of transmitted data. Some do provide full-duplex transmission and bit-oriented operation.

BISYNC has largely been replaced by the more powerful SDLC (Synchronous Data Link Control).

### LAP AND LAPB

Link Access Procedure (LAP) protocols are Data Link layer protocols for framing and transmitting data across point-to-point links. LAP was originally derived from HDLC (High-Level Data Link Control), but was later updated and renamed LAPB (LAP Balanced).

LAPB is the data link protocol for X.25. LAPB is a bit-oriented protocol derived from HDLC that ensures that frames are error free and in the right sequence. It can be used as a Data Link Layer protocol implementing the connection-mode data link service in the OSI Reference Model as defined by ITU-T Recommendation X.222.

LAPB is used to manage communication and packet framing between data terminal equipment (DTE) and the data circuit-terminating equipment (DCE) devices in the X.25 protocol stack. LAPB is essentially HDLC in Asynchronous Balanced Mode (ABM). LAPB sessions can be established by either the DTE or DCE. The station initiating the call is determined to be the primary, and the responding station is the secondary.



### Frame types

- **I-Frames (Information frames):** Carries upper-layer information and some control information. I-frame functions include sequencing, flow control, and error detection and recovery. I-frames carry send and receive sequence numbers.
- **S-Frames (Supervisory Frames):** Carries control information. S-frame functions include requesting and suspending transmissions, reporting on status, and acknowledging the receipt of I-frames. S-frames carry only receive sequence numbers.
- **U-Frames (Unnumbered Frames):** carries control information. U-frame functions include link setup and disconnection, as well as error reporting. U-frames carry no sequence numbers

### Frame format

Flag	Address	Control	Data	Checksum	Flag
01111110 (8bits)	(8bits)	(8bits)	(Variable)	(16 bits)	01111110 (8bits)

**Fig. 2.26 Frame format**

**Flag** – The value of the flag is always 0x7E. In order to ensure that the bit pattern of the frame delimiter flag does not appear in the data field of the frame (and therefore cause frame misalignment), a technique known as Bit stuffing is used by both the transmitter and the receiver.

**Address field** – In LAPB, this field has no meaning since the protocol works in a point to point mode and the DTE network address is represented in the layer 3 packets. This byte is therefore put to a different use; it separates the link commands from the responses and can have only two values: 0x01 and 0x03. 01 identifies frames containing commands from DTE to DCE and responses to these commands from DCE to DTE. 03 are used for frames containing commands from DCE to DTE and for responses from DTE to DCE.

**Control field** – it serves to identify the type of the frame. In addition, it includes sequence numbers, control features and error tracking according to the frame type.

### Modes of operation

LAPB works in the Asynchronous Balanced Mode (ABM). This mode is balanced (i.e., no master/slave relationship) and is signified by the SABM (E)/SM frame. Each station may initialize, supervise, recover from errors, and send frames at any time. The DTE and DCE are treated as equals.

**FCS** – The Frame Check Sequence enables a high level of physical error control by allowing the integrity of the transmitted frame data to be checked.

**Window size** – LAPB supports an extended window size (modulo 128 and modulo 32768) where the maximum number of outstanding frames for acknowledgment is raised from 7 (modulo 8) to 127 (modulo 128) and 32767 (modulo 32768).

### Protocol operation

LAPB has no master/slave node relationships. The sender uses the Poll bit in command frames to insist on an immediate response. In the response frame this same bit becomes the receivers Final bit. The receiver always turns on the Final bit in its response to a command from the sender with the Poll bit set. The P/F bit is generally used when either end becomes unsure about proper frame sequencing because of a possible missing acknowledgment, and it is necessary to re-establish a point of reference. It is also used to trigger an acknowledgment of outstanding I-frames.

### Protocol verification:

### Finite State Machine Models

A **finite-state machine (FSM)** or **finite-state automaton** (plural: automata), or simply a **state machine**, is a mathematical model of computation used to design both computer programs and sequential logic circuits. It is conceived as an abstract machine that can be in one of a finite number of states. The machine is in only one state at a time; the state it is in at any given time is called the current state. It can change from one state to another when initiated by a triggering event or condition; this is called a transition. A particular FSM is defined by a list of its states, and the triggering condition for each transition.

Finite-state machines can model a large number of problems, among which are electronic design automation, communication protocol design, language parsing and other engineering applications. In

biology and artificial intelligence research, state machines or hierarchies of state machines have been used to describe neurological systems. In linguistics, they are used to describe simple parts of the grammars of natural languages.

### The FSM Consist of

- **States** are those instants that the protocol machine is waiting for, the next event to happen e.g. waiting for ACK.
- **Transitions** occur when some event happens. E.g. when a frame is sent, when a frame is arriving, when timer goes off, when an interrupt occurs.
- **Initial State** gives description of the system i.e. when it starts running.
- A **deadlock** is a situation in which the protocol can make no more forward progress, there exists a set of states from which there is no exit and no progress can be made.

### How to know a protocol really works → specifies and verify protocol using, e.g. finite state machine

- Each protocol machine (sender or receiver) is at a specific state at every time instant
- Each state has zero or more possible transitions to other states
- One particular state is initial state: from initial state, some or possibly all other states may be reachable by a sequence of transitions.
- Simplex stop and wait ARQ protocol:
  - State SRC:  $S = 0, 1 \rightarrow$  which frame sender is sending;
  - $R = 0, 1 \rightarrow$  which frame receiver is expecting;
  - $C = 0, 1, A (ACK), - (empty) \rightarrow$  channel state, i.e. what is in channel

### There are 9 transitions

Transition	Who runs?	Frame Accepted	Frame Emitted	To Network Layer
0	–	Frame lost	Frame lost	
1	R	0	A	–
2	S	A	1	Yes
3	R	1	A	–
4	S	A	0	Yes
5	R	0	A	–
6	R	1	A	No
7	S	Time out	0	No
8	S	Time out	1	–

**Table 2.1 List of Transitions**

- Initial state (000): sender has just sent frame 0, receiver is expecting frame 0, and frame 0 is currently in channel
- Transition 0: States channel losing its content.
- Transition 1: consists of channel correctly delivering packet 0 to receiver, and receiver expecting frame 1 and emitting ACK 0. Also receiver delivering packet 0 to the network layer.
- During normal operation, transitions 1,2,3,4 are repeated in order over and over: in each cycle, two frames are delivered, bringing sender back to initial state.

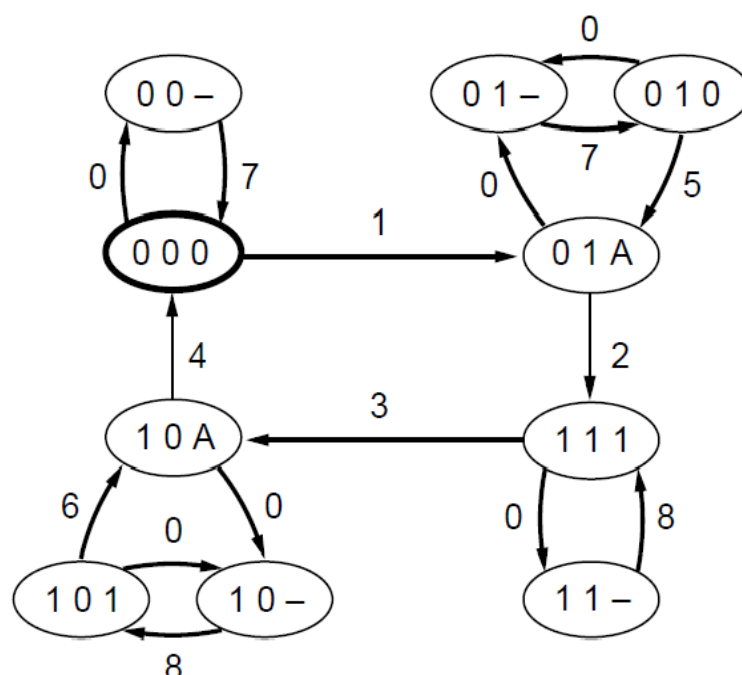


Fig 2.27 FSM for Stop and Wait Protocol (Half Duplex)

### Petri Net

Petri Net(PN) is an abstract model to show the interaction between asynchronous processes. It is only one of the many ways to represent these interactions. Asynchronous means that the designer doesn't know when the processes start and in which sequence they'll take place. A common manner to visualize the concepts is with the use of places, tokens, transitions and arcs. We refer to the basics of Petri Net for a first introduction in notations. We want to mention that a transition can only fire when there are tokens in every input-place. When it fires, one token is taken from every input-place and every output-place from the transition gets an (extra) token.

#### The Basics:

A Petri Net is a collection of directed arcs connecting places and transitions. Places may hold tokens. The state or marking of a net is its assignment of tokens to places. Here is a simple net containing all components of a Petri Net:

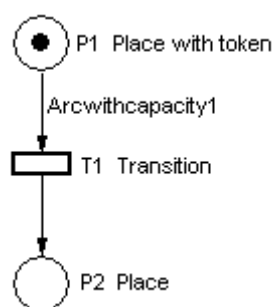
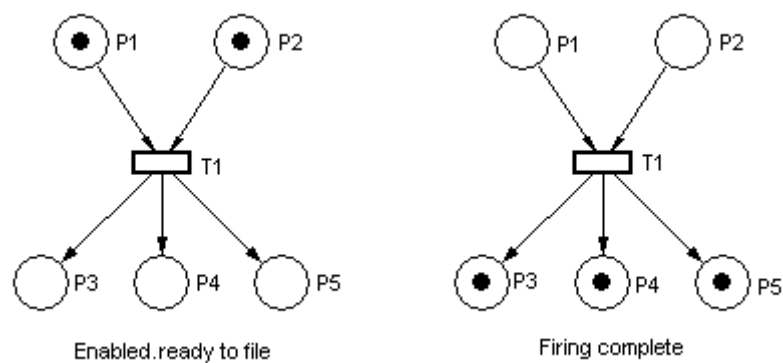


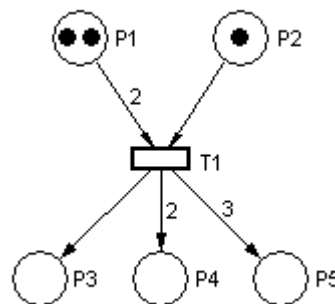
Fig 2.28 Petri Net Model

Arcs have capacity 1 by default; if other than 1, the capacity is marked on the arc. Places have infinite capacity by default, and transitions have no capacity, and cannot store tokens at all. With the rule that arcs can only connect places to transitions and vice versa, we have all we need to begin using Petri Nets. A few other features and considerations will be added as we need them.

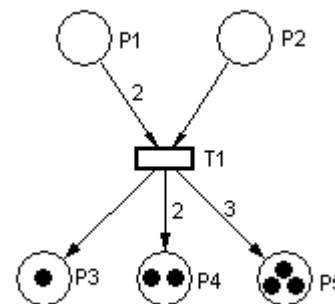
A transition is enabled when the number of tokens in each of its input places is at least equal to the arc weight going from the place to the transition. An enabled transition may fire at any time. When fired, the tokens in the input places are moved to output places, according to arc weights and place capacities. This results in a new marking of the net, a state description of all places.

**Fig 2.29**

When arcs have different weights, we have what might at first seem confusing behaviour. Here is a similar net, ready to fire:

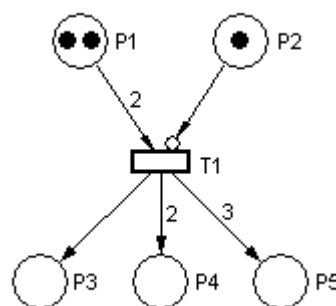
**Fig 2.30**

and here it is after firing:

**Fig 2.31**

When a transition fires, it takes the tokens that enabled it from the input places; it then distributes tokens to output places according to arc weights. If the arc weights are all the same, it appears that tokens are moved across the transition. If they differ, however, it appears that tokens may disappear or be created. That, in fact, is what happens; think of the transition as removing its enabling tokens and producing output tokens according to arc weight.

A special kind of arc, the inhibitor arc, is used to reverse the logic of an input place. With an inhibitor arc, the absence of a token in the input place enables, not the presence:

**Fig 2.32**

This transition cannot fire, because the token in P2 inhibits it.

### Tokens can play the following roles:

A physical object: a robot;

- An information object: a message between two robots;
- A collection of objects: the people mover;
- An indicator of a state: the state in which a robot is: defender/attacker;
- An indicator of a condition: a token indicates whether a certain condition is fulfilled (ex. Soccer game starts when the referee gives the signal).

### Transitions can play the following roles:

- An event: start a thread, the switching of a machine from normal to safe mode;
- A transformation of an object: a robot that changes his role, see further;
- A transport of an object: the ball is passed between the robots.

An arc connects only places and transitions and indicates the direction in which the token travels.

Petri net Link <https://www.youtube.com/watch?v=EmYVZuczJ6k>

Finite State Machine Models <https://www.youtube.com/watch?v=hJIST1cEf6A>

SDLC AND HDLC <https://www.youtube.com/watch?v=fwVTFO-u4g>

### ARP:

ARP or Address Resolution Protocol is a simple communications protocol used primarily today in IP and Ethernet networks. Its main purpose is to discover and associate IP addresses to physical MAC hardware addresses. [ARP](#) is used to find the MAC address of device on a network using only the IP address. The ARP protocol will make a broadcast out to the network asking for the MAC address of the destination IP address. The machine with the IP address will respond with its MAC address. The communication then drops to the link layer for physical to physical data communication between computers. ARP's job is to basically discover and associate IP addresses to physical MAC addresses.

### RARP:

RARP (Reverse ARP) is a legacy protocol that has now been replaced by BOOTP and later by DHCP. Its purpose was for diskless workstations (i.e no ability to store an IP address) to discover what their own IP address was - based on their MAC address. At the point of boot, the workstation would send a request requesting its IP, a RARP server would then respond with the appropriate IP. For example:

RARP Request: What is my IP address (MAC address is within Ethernet header)?

RARP Response: Your IP address is 192.168.1.11.

The main problem with RARP was that:

- The RARP server needed to be populated with the MAC to IP mappings.
- No additional data (DNS, NTP) could be sent other than the IP address.



- It only operates within a broadcast domain.

RARP was, therefore, superseded by BOOTP. However, BOOTP still required a static mapping to be defined (MAC to IP). DHCP was then built upon BOOTP with the ability to use a pool of addresses.

#### **GARP:**

In more advanced networking situations you may run across something known as Gratuitous ARP (GARP).

A gratuitous arp something that is often performed by a computer when it is first booted up. When a NIC's is first powered on, it will do what's known as a gratuitous ARP and automatically ARP out it's MAC address to the entire network. This allows any switches to know the location of the physical devices and DHCP servers to know where to send an IP address if needed and requested. Gratuitous ARP is also used by many high availability routing and load balancing devices. Routers or load balancers are often configured in an HA (high availability) pair to provide optimum reliability and maximum uptime. Usually these devices will be configured in an Active/Standby pair. One device will be active while the second will be sleeping waiting for the active device to fail. Think of it as an understudy for the lead role in a movie. If the leading lady gets sick, the understudy will gladly and quickly take her place in the lime light.

When a failure occurs, the standby device will assert itself as the new active device and issue a gratuitous ARP out to the network instructing all other devices to send traffic to it's MAC address instead of the failed device.



Computer Networks



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