

# 18CSC302J- Computer Networks Unit-2

### **BYTE ORDERING**

# Byte ordering

- An arrangement of bytes when data is transmitted over the network is called byte ordering. Different computers will use different byte ordering.
- When communication taking place between two machines byte ordering should not make discomfort.
- Generally an Internet protocol will specify a common form to allow different machines byte ordering. TCP/IP is the Internet Protocol in use.
- Two ways to store bytes: Big endian and little endian
- Big-endian High order byte is stored on starting address and low order byte is stored on next address
- Little-endian Low order byte is stored on starting address and high order byte is stored on next address

# Byte ordering functions

• Special functions are applied through routines to convert host's internal byte order representation to network byte order.

Name of the function	Description
htons()	Host to network short
htonl()	Host to network long
ntohs()	Network to host short
ntohl()	Network to host long

#### **BYTE ORDERING FUNCTIONS(cont)**

- unsigned short htons() This function converts 16-bit (2-byte) data from host byte order to network byte order.
- **unsigned long htonl()** This function converts 32-bit (4-byte) data from host byte order to network byte order.
- **unsigned short ntohs()** This function converts 16-bit (2-byte) data from network byte order to host byte order.
- **unsigned long ntohl()** This function converts 32-bit (4- byte) data from network byte order to host byte order.

### **SYSTEM CALLS AND SOCKETS**



# System calls

#### **System Calls - An interface between process and operating systems**

- It provides
  - > the services of the operating system to the user programs via Application Program Interface(API).
  - > An interface to allow user-level processes to request services of the operating system.
  - > System calls are the only entry points into the kernel system.

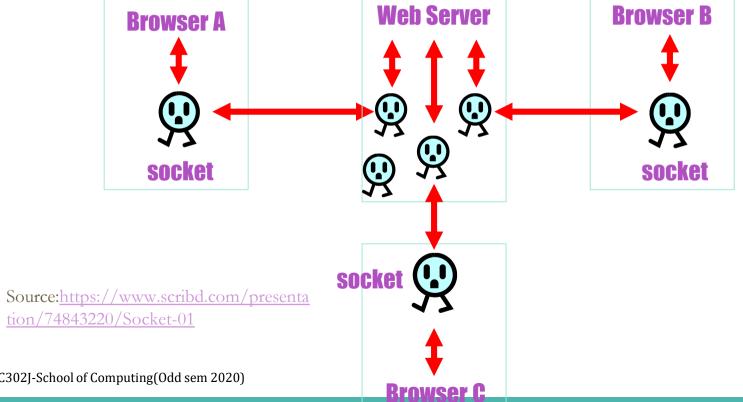
#### **Sockets**

 Socket is an interface between applications and the network services provided by operating systems

# APPLICATION SOCKET - APPICATION INTERFACE(API) TCP/IPV4, TCP/IPV6, UNIX

- Applications use sockets to send and receive the data
- Socket provides IP address and port address

#### **Socket Abstraction**



# **Socket Descriptors**

- To perform file I/O, **file descriptor** is used.
- To perform network I/O, **socket descriptor** is used
- Each active socket is identified by its socket descriptor.
- The data type of a socket descriptor is **SOCKET**.
- Hack into the header file winsock2.h:
- > typedef u\_int SOCKET; /\* u\_int is defined as unsigned int \*/
- In UNIX systems, socket is just a special file, and socket descriptors are kept in the file descriptor table.
- The Windows operating system keeps a separate table of socket descriptors (named **socket descriptor table**, or SDT) for each process.

#### **Socket creation**

• The socket API contains a function socket() that can be called to create a socket. e.g.:

```
#include <winsock2.h>
...
SOCKET s;
...
s = socket(AF_INET, SOCK_DGRAM, 0);
```

The socket is not usable yet. We need to specify more information before using it for data transmission.

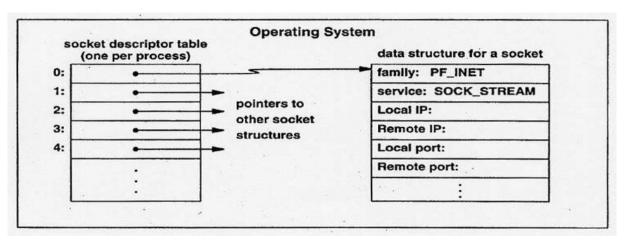
# **Types of Sockets**

- Under protocol family AF\_INET
  - Stream socket
    - Uses TCP for connection-oriented reliable communication
    - Identified by SOCK\_STREAM
    - $\blacksquare$   $s = socket(AF_INET, SOCK_STREAM, 0);$
  - Datagram socket
    - Uses UDP for connectionless communication
    - Identified by SOCK\_DGRAM
    - $s = socket(AF\_INET, SOCK\_DGRAM, 0);$
  - RAW socket
    - Uses IP directly
    - Identified by SOCK\_RAW
    - Advanced topic. Not covered by COMP2330.

# **System Data Structures for Sockets**

- When an application process calls socket(), the operating system allocates a new data structure to hold the information needed for communication, and fills in a new entry in the process's socket descriptor table (SDT) with a pointer to the data structure.
  - A process may use multiple sockets at the same time. The socket descriptor table is used to manage the sockets for this process.
  - Different processes use different SDTs.
- The internal data structure for a socket contains many fields, but the system leaves most of them unfilled. The application must make additional procedure calls to fill in the socket data structure before the socket can be used.
- The socket is used for data communication between two processes (which may locate at different machines). So the socket data structure should at least contain the address information, e.g., **IP addresses, port numbers**, etc.

## System Data Structures for Sockets (Cont.)



• Conceptual operating system (Windows) data structures after **five calls** to *socket()* by a process. The system keeps a separate socket descriptor table for each process; threads in the process share the table.

# Functions used in client program

- > socket()- create the socket descriptor
- connect()- connect to the remote server
- read(),write()- communicate with the server
- close()- end communication by closing socket descriptor

# Socket()

#### int socket(int domain, int type, int protocol)

Returns a descriptor (or handle) for the socket

- **Domain:** protocol family

  PF INET for the Internet
- Type: semantics of the communication

SOCK\_STREAM: Connection oriented

SOCK DGRAM: Connectionless

- **Protocol**: specific protocol
  - **UNSPEC**: unspecified

(PF\_INET and SOCK\_STREAM already implies TCP)

- E.g., TCP: sd = socket(PF\_INET, SOCK\_STREAM, 0);
- E.g., UDP: sd = socket(PF\_INET, SOCK\_DGRAM, 0);

# Connect()

int connect(int sockfd, struct sockaddr \*server\_address, socketlen\_t addrlen)

- Arguments: socket descriptor, server address, and address size
- Remote address and port are in struct sockaddr
- Returns 0 on success, and -1 if an error occurs

# Read(), Write() and Close()

#### **Sending data**

write(int sockfd, void \*buf, size\_t len)

- Arguments: socket descriptor, pointer to buffer of data, and length of the buffer
- Returns the number of characters written, and -1 on error

#### **Receiving data**

read(int sockfd, void \*buf, size\_t len)

- Arguments: socket descriptor, pointer to buffer to place the data, size of the buffer
- Returns the number of characters read (where 0 implies "end of file"), and -1 on error

#### Closing the socket

int close(int sockfd)

# Functions used in server program

- socket()- create the socket descriptor
- bind()- associate the local address
- listen()- wait for incoming connections from clients
- accept()- accept incoming connection
- read(),write()- communicate with client
- close()- close the socket descriptor

# Bind()

Bind socket to the local address and port

int bind (int sockfd, struct sockaddr \*my\_addr, socklen\_t addrlen)

• Arguments: socket descriptor, server address, address

length

• Returns 0 on success, and -1 if an error occurs

# Listen()

Define the number of pending connections

int listen(int sockfd, int backlog)

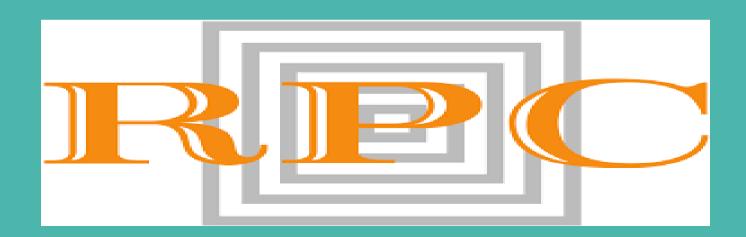
- Arguments: socket descriptor and acceptable backlog
- Returns 0 on success, and -1 on error

# Accept()

#### int accept(int sockfd, struct sockaddr \*addr, socketlen\_t \*addrlen)

- Arguments: socket descriptor, structure that will provide client address and port, and length of the structure
  - Returns descriptor for a new socket for this connection
- What happens if no clients are around?
  - § The accept() call blocks waiting for a client
- What happens if too many clients are around?
  - § Some connection requests don't get through
  - § ... But, that's okay, because the Internet makes no promises

### **RPC - REMOTE PROCEDURE CALL**



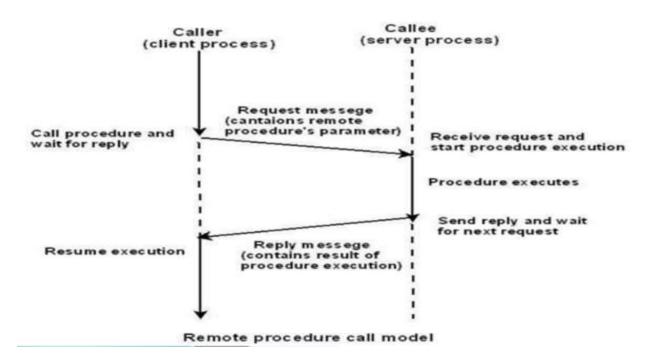


## RPC Model(Remote Procedure Call)

- A *remote procedure call* is an interprocess communication technique that is used for client-server based applications. It is also known as a subroutine call or a function call.
- RPC allows programs to call the procedure which is located on the other machines.
- Message passing is not visible to the programmer, so it is called as Remote Procedure call (RPC).
- RPC enables a procedure call that does not reside in the address space of the calling process.
- In RPC, the caller and the callee has *disjoint address* space, hence there is *no access to data and variables* in the callers environment.
- RPC performs a *message passing scheme* for information exchange between the caller and the callee process.



## RPC Model(Remote Procedure Call)



Source:https://www.slideshare.net/sunita sahu 101/rpc-remote-procedure-call

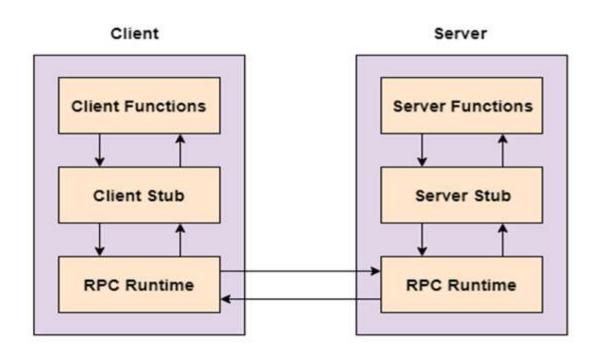


### RPC Model(Remote Procedure Call)

- A client has a request message that the RPC translates and sends to the server.
- This request may be a procedure or a function call to a remote server.
- When the server receives the request, it sends the required response back to the client.
- The client is blocked while the server is processing the call and only resumed execution after the server
  is finished.



#### **Client Server RPC Model**



Source:https://www.tutorialspoint.com/remote-procedure-call-rpc



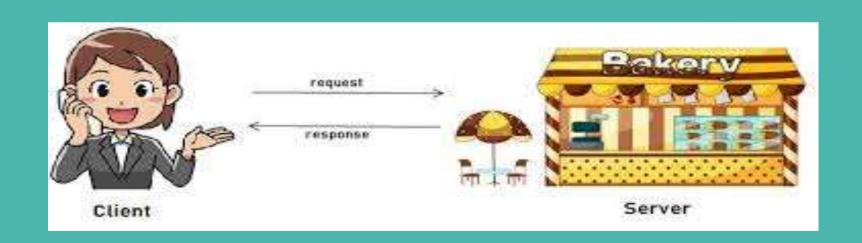
### Sequence of events in a RPC

- The client stub is called by the client.
- The client stub makes a system call to send the message to the server and puts the parameters in the message.
- The message is sent from the client to the server by the client's operating system.
- The message is passed to the server stub by the server operating system.
- The parameters are removed from the message by the server stub.
- Then, the server procedure is called by the server stub.

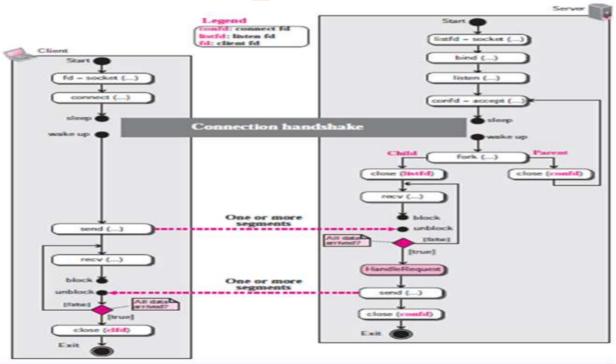
#### **RPC Features**

- Remote procedure calls support process oriented and thread oriented models.
- The internal message passing mechanism of RPC is hidden from the user.
- The effort to re-write and re-develop the code is minimum in remote procedure calls.
- Remote procedure calls can be used in distributed environment as well as the local environment.
- Many of the protocol layers are omitted by RPC to improve performance.
- Ease of use, efficiency.

## TCP CLIENT -SERVER AND PACKAGES

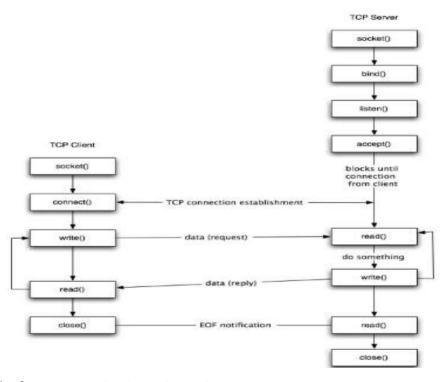


# **Client Server Program**



Source: Behrouz A. Forouzan, "TCP IP Protocol Suite" 4th edition, 2010, McGraw-Hill ISBN: 0073376043 18CSC302J- School of Computing (Odd sem 2020)

# **Client Server Program**



## **Server Processing using TCP**

- Create a socket using the socket() function in c.
- Initialize the socket address structure and bind the socket to an address using the bind() function.
- Listen for connections with the listen() function.
- Accept a connection with the accept() function system call.
- This call typically blocks until a client connects to the server.
- Receive and send data by using the recv() and send() function in c.
- Close the connection by using the close() function

## **Client Processing using TCP**

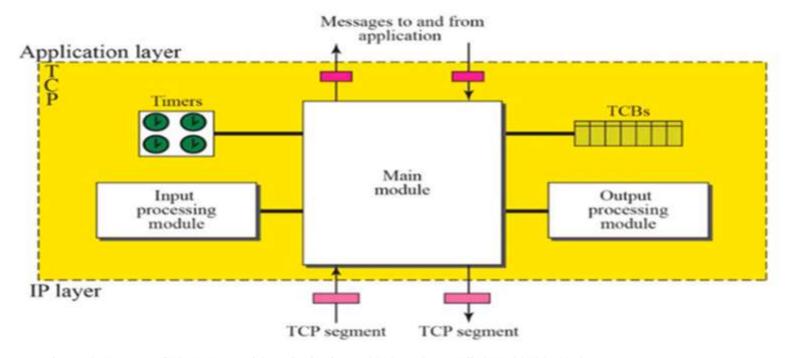
- Create a socket using the socket() function\_in c.
- Initialize the socket address structure as per the server and connect the socket to the address of the server using the connect();
- Receive and send the data using the recv() and send() functions.
- Close the connection by calling the close() function.

## **Input, Output Processing Module**

#### **TCP Package**

- TCP is a stream-service, connection-oriented protocol with an involved state transition diagram.
- It uses flow and error control.
- It is so complex because actual code includes tens of thousands of lines.

# **TCP Package Diagram**



Source: Behrouz A. Forouzan, "TCP IP Protocol Suite" 4th edition, 2010, McGraw-Hill ISBN: 0073376043

# **TCP Package**

- TCP Package involves tables called Transmission control blocks
- A set of timers
- Three software modules:
  - Main module
  - An input processing module
  - An output processing module

# **Input Processing Module**

- The Input processing module handles all the details which is required for the process of data or an Acknowledgement received when **TCP** is in the **ESTABLISHED STATE**.
- This module sends an **ACK** if needed.
- Takes care of the window size
- Performs Error checking and so on.

# **Output Processing Module**

- The output processing module handles all the details needed to send out data received from application program when TCP is in the **ESTABLISHED STATE**.
- This module handles retransmission time-outs, persistent time-outs and so on.

## UDP - USER DATAGRAM PROTOCOL





### Introduction

- Connectionless
- Unreliable transport protocol
- Located between application layer and network layer in the TCP/IP protocol suite
- Process to process communication using port numbers



### **Limitations of UDP**

- There is no flow control mechanism.
- There is no acknowledgement for received packets.
- Does no provide error control to some extent.



# Why would a process want to use UDP when it is powerless?

- It is simple protocol with minimum overhead.
- Application which use small messages to be sent without reliability then UDP is best.
- For small messages less no. of interactions is required between sender and receiver for UDP compared to TCP.



#### **UDP Services**

 Process to process communication – using sockets with the combination of IP address and port numbers

#### Connectionless services

- UDP is an independent datagram
- The user datagram is not numbered
- No difference between different user datagram even the source and destination are the same
- UDP can't send a stream of data. Hence the message should fit in one user datagram(less than 65,507 bytes)



- **Flow control** –no flow control, no window mechanism so receiver may overflow with incoming messages.
- **Congestion control** does not provide congestion control and has an assumption that the packets are small and sporadic so they cant create congestion.

#### Error control

- > no error mechanism except checksum.
- > The sender does not know about the packet lost
- ➤ If the receiver checks the error through checksum then that user datagram is discarded.

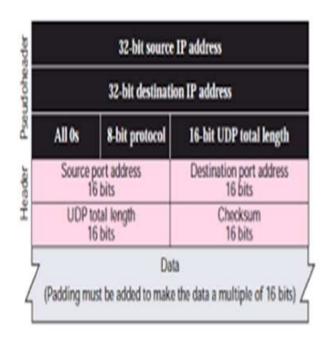


#### Checksum

contain three parts

#### 1. Pseudo header

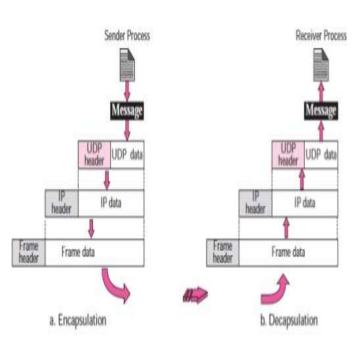
- It is a part of the header of the IP header
- Encapsulated with some fields with 0's
- Protocol field to differentiate between UDP and TCP
  - The value of the protocol field is 17. If it is changed then the packet gets discarded at receiver end.
- 2. UDP header
- 3. **Data** communicating from the application layer



Source: https://www.ques10.com/p/10930/how-is-checksum-computed-in-udp-1/



- Encapsulation and Decapsulation
  - The process sends the message to the UDP along with a pair of socket address and length of data
  - ➤ The UDP then passes to IP adding UDP header
  - The IP adds its own header along with the value 17 to indicate it is a UDP message and sends to data link layer
  - The data link layer adds its own header and passes it to physical layer
  - The physical layer encodes bits to electrical signals and sends to remote machine
- The reverse process happens on the other end for decapsulation



#### Source:

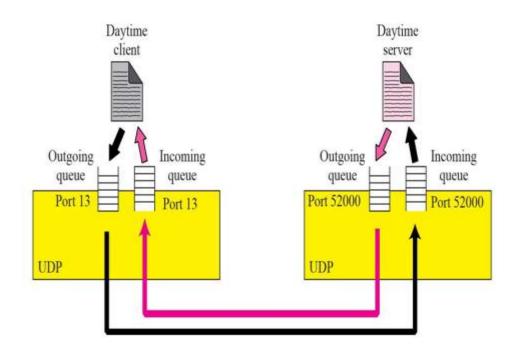
https://www.slideshare.net/MelvinCaba tuan1/transport-layer-services



#### Queuing

- Port numbers are assigned by OS
- Each process has one port number, one incoming and one outgoing queue.
- When the process terminates the queue is destroyed.
- Client side uses ephemeral port numbers
  - The client sends the message to UDP using the outgoing queue
  - If the client receives the message the UDP checks if there is a incoming queue created. If it is available it will deliver else discards the packet sending a ICMP message "port unreachable" to the server.
- Server side uses well known port numbers.
  - The server side the queue remains open as long as the server is running
  - If the server receives the message the UDP checks if there is a incoming queue created. If it is available it will deliver else discards the packet sending a ICMP message "port unreachable" to the client





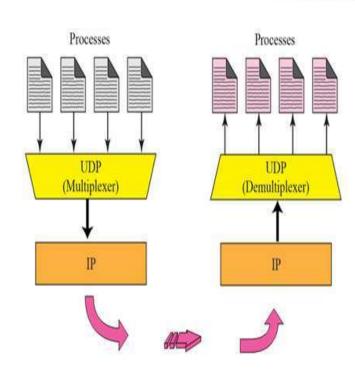
#### Source:

http://www.myreadingroom.co.in/notes-and-studymaterial/68-dcn/848-userdatagram-protocol-udp.html



#### Multiplexing

- > many to one relationship.
- UDP accepts messages from different process and differentiate by port numbers.
- Adds a header and then sends to the IPDemultiplexing
  - > One to many relationship
  - ➤ UDP receives the user datagram from IP and drops the header then sends the message to appropriate process based on port numbers



Source: https://www.rfwireless-world.com/Terminology/Advantages-and-Disadvantages-of-UDP.html



### **UDP Features**

- Connectionless services
- Lack of error control
- Lack of congestion control



### **Connectionless service**

- Preferable for small message which fits in a single datagram.
- The overhead to establish and close a connection may be significant whereas in TCP it takes 9 packets for exchanges between client and server to achieve the above goal.
- Provides less delay



#### Lack of error control

- UDP does not provide error control
- Provides unreliable service
- In reliable service the transport layer needs to take care of the lost packet by resending it. So there will be a uneven delay between different parts of the message delivered.



# **Lack of congestion control**

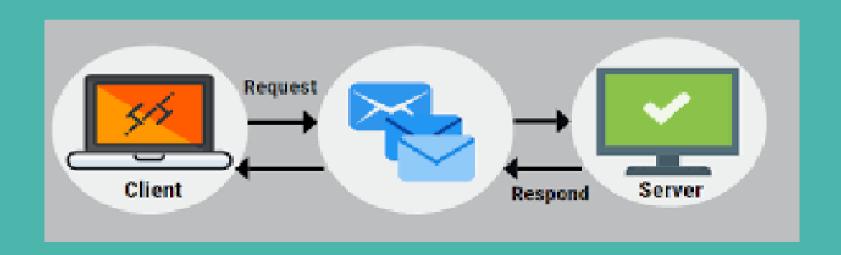
- UDP does not provide congestion control.
- UDP does not provide additional traffic in error prone network.
- TCP leads to creation of congestion or additional congestion in network by resending packets several times when a packet are lost.



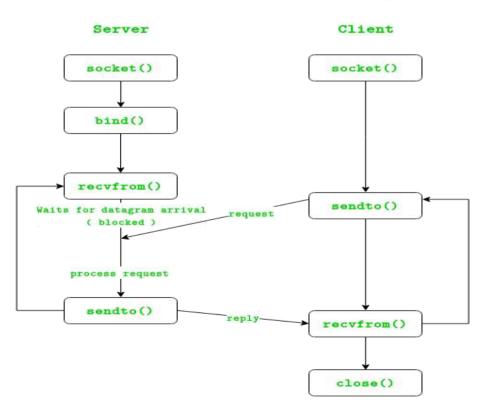
# **UDP Applications**

- Suitable for request- response communication
- Suitable for internal flow and error control mechanism
- Suitable for multicasting
- Used for management processes such as SNMP
- Used for route updating protocols such as RIP
- Used for real time applications which does not accept uneven delay

# **UDP CLIENT - SERVER AND PACKAGES**



# **Client-Server Program**



#### Source:

https://www.geeksforgeeks.org/udp-server-client-implementation-c/

# **Server Processing using UDP**

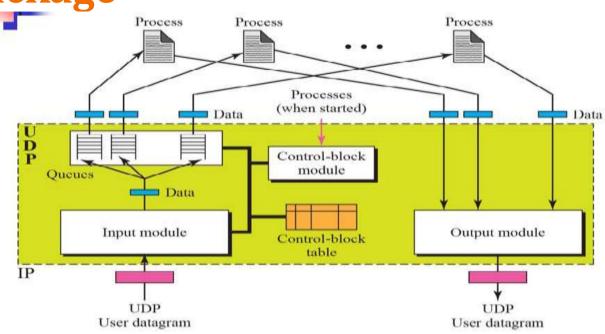
- 1. Create UDP socket.
- 2. Bind the socket to server address.
- 3. Wait until datagram packet arrives from client.
- 4. Process the datagram packet and send a reply to client.
- 5. Go back to Step 3.

# **Client Processing using UDP**

- 1. Create UDP socket.
- 2. Send message to server.
- 3. Wait until response from server is received.
- 4. Process reply and go back to step 2, if necessary.
- 5. Close socket descriptor and exit.



**UDP Package** 



Source: https://www.slideshare.net/Thanveen/user-datagram-protocol-for-msc-cs-42037663



# **Five components**

- Control-block table
- Input queues
- Control-block module
- Input module
- Output module



### **Control - block Table**

- Used to keep track of open ports
- Each entry in the table has 4 entries

- State FREE / IN-USE
- Process Id
- Port number
- Corresponding queue number



# **Input Queues**

- Each process has a set of input queue.
- We do not use output queue



### **Control- block module**

- It is responsible for management of control block table.
- When a process starts, it asks for port no from OS.
- The OS assigns well known ports for server and ephemeral ports for clients.
- The process passes the Process Id and port no to the control block module to create an entry in the table for a process.
- The value of the field queue is zero since the control-block module does not create queues.



### Control - block module contd.

UDP\_Control\_Block\_Module (process ID, port number) 2. 3. Search the table for a FREE entry. 4. if (not found) 5. Delete one entry using a predefined strategy. 6. Create a new entry with the state IN\_USE Enter the process ID and the port number. 8. Return. } //end module



# Input module

- It receives a user datagram from IP.
- It searches the control block table to find an entry having the same port number as this user datagram
  - If found -> the module uses the info in the entry to enqueue the data.

➤ If not found -> it generates an ICMP message.



# Input - module contd.

```
1. UDP_INPUT_Module(user_datagram)
2. {
3.
      Look for the entry in the control_block table
4.
      if (found)
5.
6.
            Check to see if a queue is allocated
            Allocate a queue
8.
      else
9.
            Enqueue the data
10. } //end if
11. else
12. {
13.
         Ask ICMP to send an "unreachable port" message
14.
          Discard the user datagram
15. }//end else
16. Return
17. } //end module
```



# **Output module**

It is responsible for creating and sending user datagrams

- 1. UDP\_OUTPUT\_MODULE(Data)
- 2. {
- 3. Create a user datagram
- 4. Send the user datagram
- 5. Return.
- 6.



# Stream Control Transmission Protocol (SCTP)

- > SCTP is designed as a general-purpose transport layer protocol that can handle multimedia and stream traffic, which are increasing every day on the Internet.
- ➤ It is a new reliable, message-oriented transport-layer protocol.

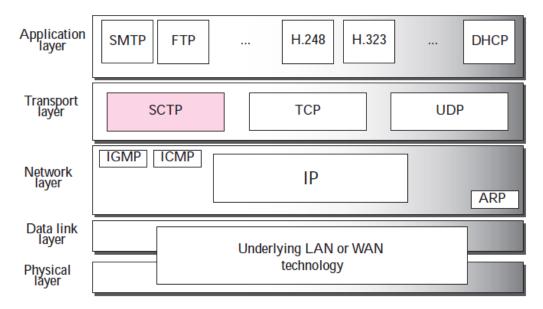


Fig: Relationship of SCTP to the other protocols in the Internet protocol suite



# Stream Control Transmission Protocol (SCTP)

Table: Comparison between UDP, TCP, and SCTP

UDP	TCP	SCTP
Message-oriented protocol	Byte-oriented protocol	Best features of UDP and TCP
UDP conserves the message boundaries	No preservation of the message boundaries	Preserves the message boundaries along with detection of lost data, duplicate data, and out-of-order data
UDP is unreliable	TCP is a reliable protocol	SCTP is a reliable message oriented Protocol
Lacks in congestion control and flow control	TCP has congestion control and flow control mechanisms	It has congestion control and flow control mechanisms

<sup>\*</sup> SCTP is a message-oriented, reliable protocol that combines the best features of UDP and TCP.

#### **SCTP Services**



#### > Process-to-Process Communication

• SCTP uses all well-known ports in the TCP space

Table : Some SCTP applications

Port Number	Description
9990	ISDN over IP
2904	SS7 telephony signaling
2905	SS7 telephony signaling
2945	Media gateway control
1718, 1719, 1720, 11720	IP telephony
5060	IP telephony
	9990 2904 2905 2945 1718, 1719, 1720, 11720

#### **➤ Multiple Streams**

- SCTP allows multistream service in each connection called as association
- If one of the streams is blocked, the other streams can still deliver their data

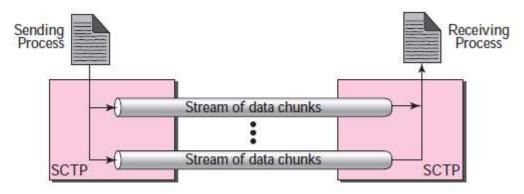


Fig: Multiple-stream concept

#### **SCTP Services**



#### > Multihoming

- The sending and receiving host can define multiple IP addresses in each end for an association.
- when one path fails, another interface can be used for data delivery without interruption.
- This fault-tolerant feature is helps on sending and receiving a real-time payload such as Internet telephony.

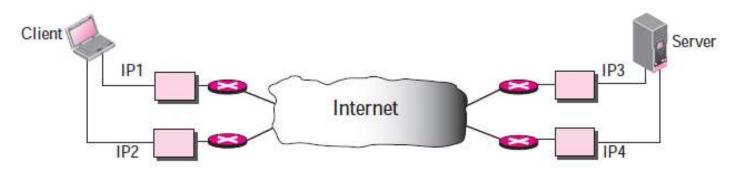


Fig: Multihoming concept

#### > Full-Duplex Communication

• SCTP has sending and receiving buffer, hence packets are sent in both directions.

# **SCTP Services**



### > Connection-Oriented Service

- A connection is called an association in SCTP
- Steps to send and receive data in SCTP
  - 1. The two SCTPs establish an association between each other.
  - 2. Data are exchanged in both directions.
  - 3. The association is terminated.

## > Reliable Service

• It uses an acknowledgment mechanism to check the safe and sound arrival of data.



## **➤ Transmission Sequence Number (TSN)**

- Unit of data in SCTP is a data chunk
- Data transfer in SCTP is controlled by numbering the data chunks
- SCTP uses a TSN to number the data chunks with 32 bits long and randomly initialized between 0 and  $2^{32}$  1
- Each data chunk carry their TSN in its header

## > Stream Identifier (SI)

- Each stream in SCTP needs to be identified using a SI
- Each data chunk carry SI in its header
- when it arrives at the destination, it is placed in order in its stream
- The SI is a 16-bit number starting from 0



## > Stream Sequence Number (SSN)

- When a data chunk arrives at the destination SCTP, it is delivered to the appropriate stream in the proper order.
- In addition to an SI, SCTP defines a SSN in each data chunk in each stream

### > Packets

- Data are carried as data chunks, control information as control chunks
- Several control chunks and data chunks can be packed together in a packet

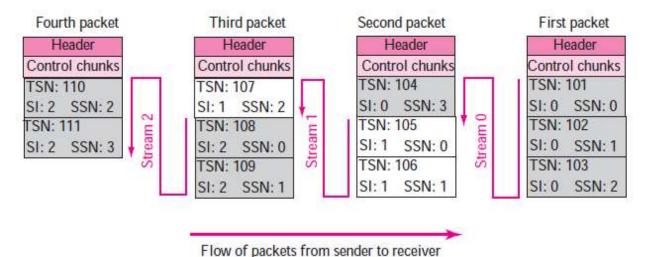
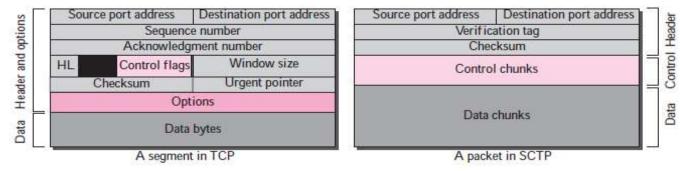


Fig: Packet, data chunks, and streams





Comparison between a TCP segment and an SCTP packet

TCP segment	SCTP packet
Control information is part of the header	Control information is included in the control chunks
Data is treated as one entity	Carry several data chunks, each can belong to a different stream
Options section exist separately	Options are handled by defining new chunk types
Mandatory part of header is 20 bytes	General header is only 12 bytes
Checksum is 16 bits	Checksum is 32 bits
Combination of IP and port addresses define a connection	Verification tag is an association identifier
Includes one sequence number in the header	Includes several different data chunks
Some segments carry control information	Control chunks never use a TSN, IS, or SSN number, they are used for data chunks only



## **►** Acknowledgment Number

- SCTP acknowledgment numbers are chunk-oriented refer to TSN
- Control information is carried by control chunks, which do not need a TSN
- Control chunks are acknowledged by another control chunk of the appropriate type

### > Flow Control

• SCTP implements flow control to avoid overwhelming the receiver

### > Error Control

• TSN numbers and acknowledgment numbers are used for error control.

## **➤** Congestion Control

• SCTP implements congestion control to determine how many data chunks can be injected into the network





- ➤ Main Parts are
  - General header
  - Chunks set of blocks
- > Types of chunks
  - Control chunks controls and maintains the association
  - Data chunks carries user data

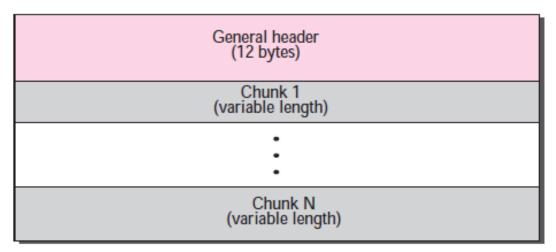


Fig: SCTP packet format



## **≻** General Header

- Defines the end points of each association to which the packet belongs
- Guarantees for a packet belongs to a particular association
- Preserves the integrity of the contents of the packet
- There are four fields in the general header
- Source port address: 16-bit field defines the port number of the sender process
- **Destination port address:** 16-bit field defines the port number of the receiving process
- Verification tag: Number that matches a packet to an association
  - It serves as an identifier for the association
  - Separate verification used for each direction in the association.
- Checksum: 32-bit field contains a CRC-32 checksum

Source port address 16 bits	Destination port address 16 bits
Verification tag 32 bits	
Checksum 32 bits	



## > Chunks

- Control information or user data are carried
- First three fields are common to all chunks
  - Type: 8-bit field define up to 256 types of chunks(few have been defined, rest are reserved for future use)
  - Flag: 8-bit field defines special flags that a particular chunk may need.
  - Length: 16-bit field defines the total size of the chunk, in bytes, including the type, flag, and length fields Table: Types of Chunks

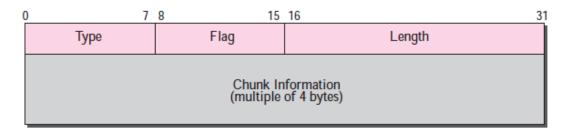


Fig: Common layout of a chunk

- Information field depends on the type of chunk
- Chunk Description Type DATA User data INIT Sets up an association INIT ACK Acknowledges INIT chunk Selective acknowledgment SACK HEARTBEAT Probes the peer for liveliness HEARTBEAT ACK Acknowledges HEARTBEAT chunk ABORT Abort an association SHUTDOWN Terminates an association Acknowledges SHUTDOWN chunk SHUTDOWN ACK Reports errors without shutting down **ERROR** Third packet in association establishment COOKIE ECHO Acknowledges COOKIE ECHO chunk COOKIE ACK Third packet in association termination SHUTDOWN COMPLETE FORWARD TSN For adjusting cumulating TSN
- SCTP requires the information section to be a multiple of 4 bytes
  - If not, padding bytes (eight 0s) are added at the end of the section



#### > Data

- Carries the user data
- A packet may contain zero or more data chunks
- Common fields
  - Type field has a value of 0
  - Flag field has 5 reserved bits and 3 defined bits
    - U signals unordered data
    - B beginning bit of fragmented message
    - E end bit of fragmented message

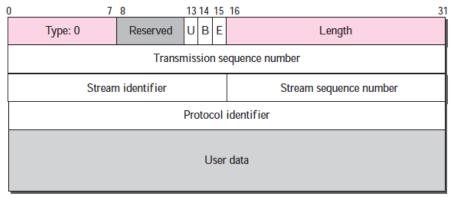


Fig: DATA chunk

- TSN Sequence number initialized in an INIT chunk for one direction and in the INIT ACK chunk for the opposite direction
- SI all chunks of same stream in one direction have same stream identifier
- Protocol identifier: 32-bit field used by the application program to define the type of data which is ignored by SCTP
- User data: carries the actual user data
  - No chunk can carry data belonging to more than one message
  - A message can be spread over several data chunks
  - Must have at least one byte of user data, can't be empty
  - If the data cannot end at a 32-bit boundary, padding must be added



## > *INIT* (Initiation chunk)

- First chunk sent by an end point to establish an association
- Cannot carry any other control or data chunks
- Verification tag for this packet is 0
- Common fields
  - Type field has a value of 1
  - Flag field is 0
  - Length field value is minimum of 20

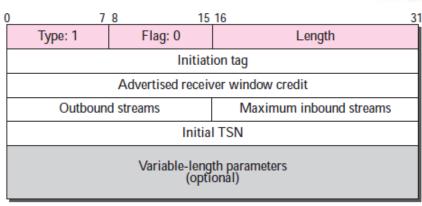


Fig: INIT chunk

- Initiation tag
  - 32-bit field defines the value of the verification tag for packets traveling in the opposite direction
  - Tag is same for all packets traveling in one direction in an association
  - Random number between 0 and  $2^{32} 1$
- Advertised receiver window credit:
  - 32-bit field used in flow control
  - Defines the initial amount of data in bytes that the sender of the INIT chunk can allow
- Outbound stream: 16-bit field defines the number of streams an initiator of the association suggests for streams in outbound direction.
- Maximum inbound stream: 16-bit field defines the maximum number of streams an initiator of the association can support in inbound direction
- Initial TSN: initializes TSN in the outbound direction
- Variable-length parameters: optional parameters to define the IP address of sending end point, multihome, cookie state etc.



- > *INIT ack*(initiation acknowledgment chunk)
- Second chunk sent during association establishment
- Value of the verification tag is the value of the initiation tag of INIT chunk.
- The parameter of type 7 defines the state cookie sent by the sender of this chunk
- Initiation tag field in this chunk initiates the value of the verification tag for future packets traveling from the opposite direction

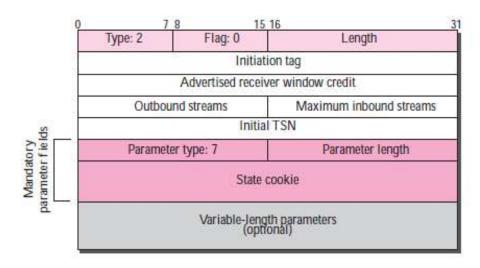


Fig: INIT ACK chunk



### > Cookie echo

- Third chunk sent during association establishment that carry user data too
- Sent by the end point that receives an INIT ACK chunk
- Chunk of type 10

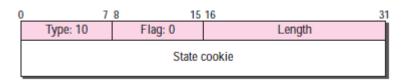


Fig: COOKIE ECHO chunk

### > COOKIE ACK

- fourth and last chunk sent during association establishment with data chunk too
- sent by an end point that receives a COOKIE ECHO chunk
- chunk of type 11

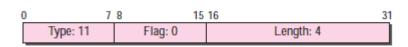


Fig: COOKIE ACK



- > *SACK*(selective ACK chunk)
- Acknowledges the receipt of data packets
- Common fields
  - Type field has 3
  - Flag bits are set to 0s

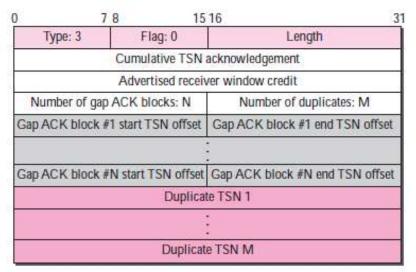


Fig: SACK chunk

- Cumulative tsn acknowledgment: 32-bit field defines the tsn of the last data chunk received in sequence
- Advertised receiver window credit: 32-bit field that have updated value for the receiver window size
- Number of gap ACK blocks: 16-bit field defines the number of gaps in the data chunk received after the cumulative TSN
- Number of duplicates: 16-bit field defines the number of duplicate chunks following the cumulative TSN
- Gap ACK block start offset: 16-bit field gives the starting TSN relative to the cumulative TSN
- Gap ACK block end offset: 16-bit field gives the ending TSN relative to the cumulative TSN
- Duplicate tsn: 32-bit field gives the tsn of the duplicate chunk.



#### > HEARTBEAT and HEARTBEAT ACK

- First has a type of 4 and the second a type of 5
- Used to periodically probe the condition of an association
- An end point sends a HEARTBEAT chunk, peer responds HEARTBEAT ACK if it is alive
- Parameter fields provide sender-specific information like address and local time
- Same is copied into the HEARTBEAT ACK chunk.

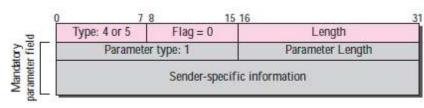


Fig: HEARTBEAT and HEARTBEAT ACK chunks

## > SHUTDOWN, SHUTDOWN ACK, and SHUTDOWN COMPLETE

- Used for closing an association
- Shutdown
  - Type 7 is eight bytes in length
  - Second four bytes define the cumulative TSN
- SHUTDOWN ACK: type 8 is four bytes in length.
- Shutdown complete
  - Type 14 is 4 bytes long
  - T flag is 1 bit flag shows that the sender does not have a TCB table

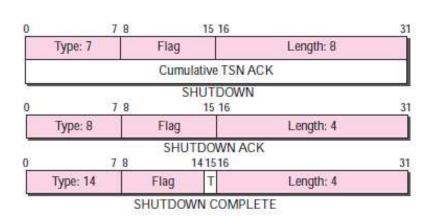


Fig: SHUTDOWN, SHUTDOWN ACK, and SHUTDOWN COMPLETE chunks



### > ERROR

- Sent when an end point finds some error in a received packet.
- It does not imply the aborting of the association.

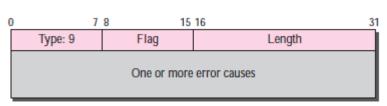


Fig: ERROR chunk

#### Description CodeInvalid stream identifier Missing mandatory parameter 3 State cookie error 4 Out of resource 5 Unresolvable address 6 Unrecognized chunk type Invalid mandatory parameters Unrecognized parameter No user data Cookie received while shutting down 10

Table: Errors

## >ABORT

• Sent when an end point finds a fatal error and needs to abort the association.

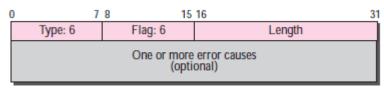


Fig: ABORT chunk

## > FORWARD TSN

• This is a chunk recently added to the standard to inform the receiver to adjust its cumulative TSN



## > Association Establishment

- Four-way handshake
  - 1. First packet has INIT chunk sent by client
    - Verification tag is 0
    - Rwnd is advertised in a SACK chunk
    - Inclusion of a DATA chunk in the third and fourth packets
  - 2. Second packet has INIT ACK chunk sent by server
    - Verification tag is the initial tag field in the INIT chunk
    - Initiates the tag to be used in the other direction
    - Defines the initial TSN and sets the servers' rwnd
  - 3. Third packet has COOKIE ECHO chunk sent by client
    - Echoes the cookie sent by the server
    - Data chunks are included in this packet
  - 4. Fourth packet has COOKIE ACK chunk sent by server
    - Acknowledges the receipt of the COOKIE ECHO chunk
    - Data chunks are included with this packet.

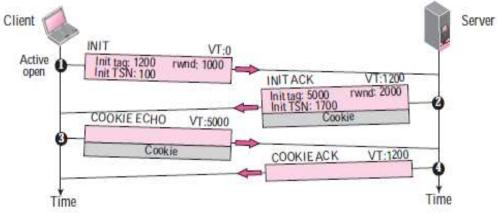


Fig: Four-way handshaking



## ➤ Number of Packets Exchanged

- Number of packets exchanged is four(3 for TCP)
- Allows the exchange of data in the third and fourth packets, so efficient

## > Verification tag

- It is a common value carried in all packets traveling in one direction in an association
- Blind attacker cannot inject a random packet into an association
- A packet from an old association cannot show up in an incarnation

## > Cookie

- Cookie is sent with the second packet to the address received in the first packet
- If the sender of the first packet is an attacker, the server never receives the third packet
- If the sender of the first packet is an honest client, it receives the second packet, with the cookie

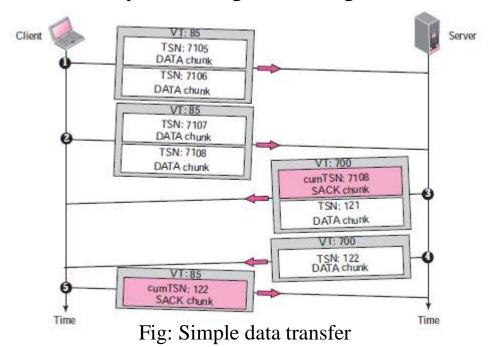


## **>** Data transfer

- Purpose of an association is to transfer data between two ends
- Once association is established, bidirectional data transfer can take place
- SCTP supports piggybacking
- Each message coming from the process is treated as one unit and inserted into a DATA chunk

• Each DATA chunk formed by a message or a fragment has one TSN and acknowledged by SACK

chunks





## ➤ Multihoming Data Transfer

- Allows both ends to define multiple IP addresses for communication
- One address is **primary address**, rest are alternative addresses
- The primary address is defined during association establishment
- Primary address of the destination is used by default for data transfer, if it is not available, one of the alternative addresses is used
- SACK is sent to the address from which the corresponding SCTP packet originated

## > Multistream delivery

- TSN numbers are used to handle data transfer whereas delivery of the data chunks are controlled by SIs and SSNs
- Two types of data delivery in each stream
  - Ordered: SSNs define the order of data chunks in the stream
  - Unordered: U flag is set, it delivers the message carrying the chunk to the destination application without waiting for the other messages



## > Fragmentation

- SCTP preserves the boundaries of the message when creating DATA chunk from a message
- If the total size exceeds the MTU, the message needs to be fragmented
- Steps for fragmentation
  - Message is broken into smaller fragments to meet the size requirement
  - DATA chunk header is added to each fragment that carries a different TSN
  - All header chunks carry the same SI, SSN, payload protocol identifier and U flag
  - B and E are assigned as
    - A. First fragment: 10
    - B. Middle fragments: 00
    - C. Last fragment: 01
- Fragments are reassembled at the destination



## **➤** Association Termination (Graceful termination)

- Either client or server involved in exchanging data can close the connection
- SCTP does not allow a "halfclosed" association, i.e. if one end closes the association, the other end must stop sending new data
- If not, the data in the queue are sent and the association is closed
- Association termination uses three packets
  - SHUTDOWN
  - SHUTDOWN ACK
  - SHUTDOWN COMPLETE

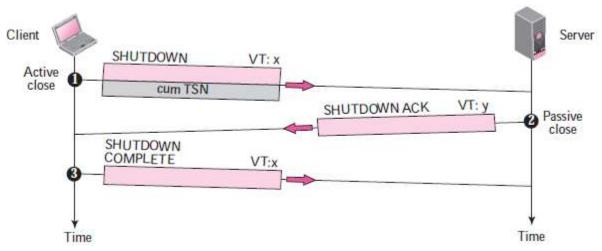


Fig: Association termination



### > Association abortion

- Association in SCTP can be aborted based requested by the process at either end or by SCTP
- A process may wish to abort the association if the process receives wrong data from the other end, going into an infinite loop etc.
- Server may wish to abort since it has received an INIT chunk with wrong parameters, requested resources are not available after receiving the cookie, the operating system needs to shut down etc.
- For abortion process either end can send an abort chunk to abort the association

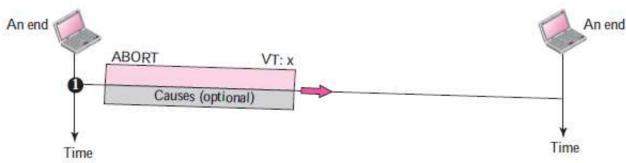


Fig: Association abortion



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