Wireless TCP and Its Variants

1. Introduction to TCP in Wireless Networks

- TCP (Transmission Control Protocol) is a connection-oriented protocol that ensures reliable data delivery over networks.
- In wireless networks, TCP faces challenges due to factors like packet loss, mobility, and varying bandwidth.
- Traditional TCP assumes packet loss is due to congestion, but in wireless networks, packet loss can also occur due to channel errors, handoffs, or interference.

2. Challenges of TCP in Wireless Networks

- **High Bit Error Rate (BER):** Wireless channels are prone to errors, leading to packet loss.
- Mobility: Handoffs between base stations or access points can cause packet loss or delays.
- **Bandwidth Fluctuations:** Wireless networks often experience varying bandwidth due to interference or shared medium.
- Congestion Misinterpretation: TCP misinterprets wireless packet loss as congestion, leading to unnecessary congestion control measures.

3. Wireless TCP Variants

To address the challenges, several TCP variants have been proposed for wireless networks:

- **TCP Reno:** The standard TCP variant that uses congestion avoidance and fast retransmit mechanisms. It performs poorly in wireless environments.
- TCP Vegas: Focuses on congestion avoidance by estimating round-trip time (RTT) and adjusting the window size accordingly. It is more proactive but still struggles with wireless-specific issues.
- TCP NewReno: An improvement over TCP Reno, it handles multiple packet losses better but still assumes packet loss is due to congestion.
- TCP SACK (Selective Acknowledgment): Allows the receiver to acknowledge non-contiguous data blocks, improving performance in lossy environments.
- **TCP Westwood:** Estimates available bandwidth and adjusts the congestion window accordingly. It is more suitable for wireless networks.
- **TCP CUBIC:** A high-speed TCP variant designed for modern networks, including wireless. It uses a cubic function to manage the congestion window.

• Explicit Congestion Notification (ECN): Allows routers to notify endpoints of congestion before packet loss occurs.

Wireless TCP Variants:

- TCP FACK (Forward Acknowledgment): Focuses on recovering from multiple packet losses.
- o **TCP Hybla:** Designed for high-latency networks, such as satellite links.
- TCP Veno: Combines TCP Vegas and TCP Reno to distinguish between congestion and wireless losses.
- TCP Jersey: Uses delay-based congestion detection to improve performance in wireless networks.

4. Key Concepts in Wireless TCP

Split-Connection Approaches:

- The connection is split into two segments: wired and wireless.
- Example: Indirect TCP (I-TCP) splits the connection at the base station, isolating wireless losses from the wired network.

• End-to-End Approaches:

- Modifications are made to the TCP protocol to handle wireless losses without splitting the connection.
- Example: Snoop Protocol caches packets at the base station and retransmits them locally in case of loss.

Cross-Layer Optimization:

- Involves collaboration between TCP and lower layers (e.g., MAC layer) to improve performance.
- Example: Explicit Loss Notification (ELN) informs TCP about wireless losses to avoid unnecessary congestion control.

5. Performance Metrics for Wireless TCP

- Throughput: The amount of data successfully transmitted per unit time.
- Fairness: Ensuring all users get a fair share of the network resources.
- Latency: The time taken for a packet to travel from sender to receiver.
- Packet Loss Rate: The percentage of packets lost during transmission.
- **Energy Efficiency:** Important for battery-powered devices in wireless networks.

Hop-by-Hop Congestion Control

1. Introduction to Hop-by-Hop Congestion Control

- **Congestion Control** is a mechanism used in networks to prevent congestion and ensure efficient data flow.
- **Hop-by-Hop Congestion Control** is a decentralized approach where each node (router or switch) in the network is responsible for managing congestion locally.
- Unlike end-to-end congestion control (e.g., TCP), which relies on feedback from the receiver, hop-by-hop control addresses congestion at every intermediate node.

2. Why Hop-by-Hop Congestion Control?

- **End-to-End Limitations:** In large or high-speed networks, end-to-end congestion control (like TCP) may not react quickly enough to prevent congestion.
- Scalability: Hop-by-hop control is more scalable for large networks with multiple hops.
- **Localized Response:** Congestion can be detected and mitigated at the source (the congested node), reducing the impact on the entire network.

3. Key Concepts in Hop-by-Hop Congestion Control

- Local Decision Making: Each node monitors its own buffer occupancy and link utilization to detect congestion.
- **Feedback Mechanisms:** Nodes communicate congestion information to their neighbors using explicit signals or implicit indicators (e.g., packet drops).
- Rate Adjustment: Nodes adjust their transmission rates based on local congestion information.

4. Techniques for Hop-by-Hop Congestion Control

- Explicit Congestion Notification (ECN):
 - Routers mark packets to indicate congestion, and the receiver sends this information back to the sender.
 - Example: ECN in IP networks.

Backpressure Mechanism:

 A congested node signals upstream nodes to reduce their transmission rates. Example: Used in data center networks and ATM networks.

• Queue Management Algorithms:

 Algorithms like RED (Random Early Detection) or CoDel (Controlled Delay) proactively manage queue lengths to prevent congestion.

Credit-Based Flow Control:

- Nodes exchange credits to control the amount of data transmitted, ensuring no node is overwhelmed.
- Example: Used in InfiniBand and some high-performance computing networks.

5. Advantages of Hop-by-Hop Congestion Control

- **Faster Reaction:** Congestion is addressed locally, reducing the time taken to mitigate it.
- Improved Fairness: Resources are allocated more fairly among flows.
- Scalability: Suitable for large networks with many hops.
- Reduced Packet Loss: Proactive congestion management minimizes packet drops.

6. Disadvantages of Hop-by-Hop Congestion Control

- Complexity: Requires coordination between nodes, increasing implementation complexity.
- Overhead: Additional signaling and processing are needed at each node.
- **Potential for Oscillations:** Poorly designed algorithms can cause oscillations in transmission rates.

7. Examples of Hop-by-Hop Congestion Control

• Data Center Networks:

Techniques like DCQCN (Data Center Quantized Congestion Notification)
use hop-by-hop control to manage congestion in large data centers.

Wireless Sensor Networks:

 Nodes adjust their transmission rates based on local congestion to conserve energy and prevent packet loss.

ATM Networks:

 ATM uses hop-by-hop flow control mechanisms like ABR (Available Bit Rate) to manage congestion.

Rate-Based Congestion Control

1. Introduction to Rate-Based Congestion Control

- Congestion Control is a mechanism used to prevent network congestion and ensure efficient data flow.
- Rate-Based Congestion Control focuses on regulating the transmission rate of data sources to match the available network capacity.
- Unlike window-based congestion control (e.g., TCP), which adjusts the number of packets in flight, rate-based control directly adjusts the sending rate.

2. Why Rate-Based Congestion Control?

- High-Speed Networks: Rate-based control is more suitable for high-speed networks where window-based mechanisms may not react quickly enough.
- **Real-Time Applications:** It is ideal for real-time applications (e.g., video streaming, VoIP) that require consistent data rates.
- **Fairness:** Rate-based control can ensure fair bandwidth allocation among multiple flows.

3. Key Concepts in Rate-Based Congestion Control

- Transmission Rate: The rate at which a sender transmits data, typically measured in bits per second (bps).
- **Feedback Mechanism:** Receivers or intermediate nodes provide feedback about network conditions (e.g., congestion, available bandwidth).
- Rate Adjustment: The sender adjusts its transmission rate based on feedback to avoid congestion.

4. Techniques for Rate-Based Congestion Control

- Explicit Rate Feedback:
 - Intermediate nodes (e.g., routers) explicitly inform the sender about the allowed transmission rate.
 - Example: ERM (Explicit Rate Marking) in ATM networks.
- Additive Increase Multiplicative Decrease (AIMD):
 - The sender gradually increases its rate (additive increase) and reduces it sharply (multiplicative decrease) in response to congestion.
 - Example: Used in some variants of TCP.

Equation-Based Rate Control:

- The sender calculates the transmission rate using a mathematical model based on network conditions.
- Example: TFRC (TCP-Friendly Rate Control).

Credit-Based Flow Control:

- The sender transmits data based on credits received from the receiver or intermediate nodes.
- Example: Used in InfiniBand and some high-performance networks.

5. Advantages of Rate-Based Congestion Control

- **Smooth Rate Adjustment:** Provides more stable and predictable data rates, suitable for real-time applications.
- Scalability: Works well in high-speed and large-scale networks.
- Fairness: Ensures fair bandwidth allocation among competing flows.

6. Disadvantages of Rate-Based Congestion Control

- Complexity: Requires accurate feedback and rate calculation mechanisms.
- **Overhead:** Additional signaling and processing are needed to provide feedback and adjust rates.
- Latency: Feedback delays can affect the responsiveness of rate adjustments.

7. Examples of Rate-Based Congestion Control

• TFRC (TCP-Friendly Rate Control):

 A rate-based protocol designed for real-time applications. It adjusts the transmission rate based on a mathematical model to compete fairly with TCP flows.

ATM ABR (Available Bit Rate):

 Uses explicit rate feedback to dynamically adjust the transmission rate based on network conditions.

• DCTCP (Data Center TCP):

 A variant of TCP used in data centers that combines rate-based and window-based control for efficient congestion management.

• QUIC (Quick UDP Internet Connections):

 A modern transport protocol that uses rate-based control for efficient congestion management in real-time applications.

Quality of Service (QoS) in Wireless Networks

1. Introduction to QoS in Wireless Networks

- Quality of Service (QoS) refers to the ability of a network to provide differentiated services to various types of traffic, ensuring that critical applications receive the necessary resources (e.g., bandwidth, latency, jitter, and packet loss).
- In wireless networks, providing QoS is challenging due to factors like limited bandwidth, interference, mobility, and varying channel conditions.

2. Why is QoS Important in Wireless Networks?

- **Diverse Traffic Types:** Wireless networks carry a mix of traffic, including voice, video, and data, each with different requirements.
- Resource Constraints: Wireless networks have limited bandwidth and are prone to interference, making efficient resource allocation critical.
- **User Expectations:** Users expect reliable and high-quality service, especially for real-time applications like VoIP and video streaming.

3. Key QoS Parameters

- **Bandwidth:** The amount of data that can be transmitted per unit time.
- Latency: The time taken for a packet to travel from source to destination.
- **Jitter:** The variation in latency between packets.
- Packet Loss: The percentage of packets lost during transmission.
- Reliability: The ability of the network to deliver packets without errors.

4. QoS Mechanisms in Wireless Networks

Traffic Prioritization:

- Assigns priority levels to different types of traffic (e.g., voice over data).
- Example: **IEEE 802.11e** for Wi-Fi networks.

Admission Control:

 Determines whether a new flow can be admitted into the network without degrading the QoS of existing flows.

Resource Reservation:

- Reserves network resources (e.g., bandwidth) for specific flows.
- Example: RSVP (Resource Reservation Protocol).
- Scheduling Algorithms:

- Determines the order in which packets are transmitted.
- Examples: Weighted Fair Queuing (WFQ), Priority Queuing.

• Congestion Management:

- Prevents network congestion by regulating traffic flow.
- Example: Random Early Detection (RED).

Error Control:

 Uses techniques like Forward Error Correction (FEC) and Automatic Repeat Request (ARQ) to reduce packet loss.

5. QoS in Specific Wireless Technologies

- Wi-Fi (IEEE 802.11):
 - IEEE 802.11e introduces QoS support through Enhanced Distributed Channel Access (EDCA), which prioritizes traffic into different access categories (e.g., voice, video, best effort, background).
- Cellular Networks (4G/5G):
 - 4G LTE uses QoS Class Identifiers (QCIs) to differentiate traffic types.
 - 5G introduces more granular QoS control with 5G QoS Identifiers (5QIs) and supports network slicing for customized QoS.

Bluetooth:

 Provides QoS through Synchronous Connection-Oriented (SCO) links for voice traffic and Advanced Audio Distribution Profile (A2DP) for high-quality audio streaming.

6. Challenges in Providing QoS in Wireless Networks

- Mobility: Users moving between cells or access points can experience service disruptions.
- **Interference:** Wireless networks are prone to interference from other devices and networks.
- **Limited Bandwidth:** Wireless networks have less bandwidth compared to wired networks, making resource allocation critical.
- **Dynamic Channel Conditions:** Wireless channels can vary rapidly due to factors like fading and multipath propagation.

(Emerging Technologies)

Bluetooth, RFID, WiMAX, Mobile IP, VoIP, and SIP

1. Bluetooth

Overview:

- Bluetooth is a wireless technology standard for short-range communication (typically up to 10 meters).
- Operates in the 2.4 GHz ISM band and uses frequency hopping to avoid interference.

Key Features:

- Low power consumption.
- Supports data and voice communication.
- Used in devices like headphones, speakers, keyboards, and IoT devices.

Protocol Stack:

- o **RF (Radio Frequency):** Handles wireless transmission.
- Baseband: Manages physical links and packet handling.
- L2CAP (Logical Link Control and Adaptation Protocol): Provides multiplexing and segmentation.
- Profiles: Define specific use cases (e.g., A2DP for audio streaming, HFP for hands-free calls).

Applications:

- Wireless audio streaming.
- File transfer between devices.
- IoT device connectivity.

2. RFID (Radio Frequency Identification)

Overview:

- RFID uses radio waves to identify and track objects.
- o Consists of tags (attached to objects) and readers (to read tag data).

Key Features:

- Passive Tags: Powered by the reader's signal.
- o Active Tags: Have their own power source.
- Operates in LF (Low Frequency), HF (High Frequency), and UHF (Ultra-High Frequency) bands.

Applications:

- Inventory management.
- o Access control (e.g., keycards).
- Supply chain tracking.

3. WiMAX (Worldwide Interoperability for Microwave Access)

Overview:

- WiMAX is a wireless broadband technology based on the IEEE 802.16 standard.
 - Provides high-speed internet access over long distances (up to 50 km).

Key Features:

- Supports both <u>fixed</u> and <u>mobile</u> broadband access.
- Uses OFDM (Orthogonal Frequency Division Multiplexing) for efficient spectrum usage.
- Offers high data rates (up to 1 Gbps).

Applications:

- Last-mile internet access in rural areas.
- Backhaul for cellular networks.
- Mobile broadband for users on the go.

4. Mobile IP

Overview:

Mobile IP is a protocol that allows devices to maintain the same IP address while moving across different networks.

Key Components:

- Mobile Node (MN): The device that moves between networks.
- Home Agent (HA): A router in the home network that forwards packets to the MN.
- o Foreign Agent (FA): A router in the visited network that assists the MN.

• How It Works:

- When the MN moves to a foreign network, it registers its new location with the HA.
- The HA tunnels packets to the MN's current location.

Applications:

- Seamless internet access for mobile devices.
- Support for roaming in wireless networks.

. VoIP (Voice over Internet Protocol)

Overview:

- VoIP enables voice communication over IP networks.
- Converts analog voice signals into digital packets for transmission.

Key Features:

- Uses protocols like RTP (Real-Time Transport Protocol) for packet delivery and SIP (Session Initiation Protocol) for call setup.
- o Supports features like call forwarding, conferencing, and voicemail

Advantages:

- Cost-effective compared to traditional telephony.
- Integrates with other IP-based services (e.g., video calls).

Applications:

- Internet-based phone calls (e.g., Skype, Zoom).
- Business communication systems (e.g., IP PBX).

6. SIP (Session Initiation Protocol)

Overview:

 SIP is a signaling protocol used to establish, modify, and terminate multimedia sessions (e.g., voice and video calls).

Key Features:

- Text-based protocol (similar to HTTP).
- Works with other protocols <u>like RTP and SDP</u> (Session Description Protocol).
- o Supports features like call transfer, conferencing, and presence.

How It Works:

- A SIP client sends an INVITE message to initiate a session.
- The recipient responds with a **200 OK** message to accept the session.

Applications:

- VolP systems.
- Video conferencing.
- Instant messaging.