StreamVerse: A Comprehensive Multi-Client
Streaming Solution for Academic Environments



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Introduction

- Multi-client streaming is a type of multimedia streaming that allows multiple users to access and view the same media content simultaneously.
- Multi-client streaming is widely used in applications such as live video streaming, web conferencing, and online training platforms.
- Multi-client streaming eliminates the limitations of the traditional clientserver model, allowing for seamless streaming to multiple clients without overloading the server.
- The system enables the efficient delivery of high-quality streaming content to a large number of clients with low latency and fast response times.

Motivation

High-quality Streaming

Multi-client streaming is a rapidly evolving technology that enables the efficient delivery of high-quality streaming content to a large number of clients with low latency and fast response times.

Traditional client-server model

The traditional client-server model has limitations in handling multiple client connections simultaneously, and multi-client streaming addresses these limitations by using various technologies and protocols.

Multiple video feeds Simultaneously

Multi-client streaming also enables the ability to view multiple video feeds simultaneously, which is particularly useful for live event coverage.

With multi-client streaming, viewers can choose which event they want to watch, and the streaming server can deliver multiple live video feeds simultaneously.

Motivation

Cross Platforms

Multi-client streaming gives the flexibility in terms of the devices and platforms that can access the stream, making it accessible to a wide range of users.

Reduce Individual bandwidth consumption

Multi-client streaming allows for scalable and efficient distribution of video content to a large number of users, without the need for individual streaming connections for each user.

Enhanced Feature

It allows for real-time feedback and interaction between viewers, which can facilitate collaboration and knowledge sharing and can be useful and different fields.

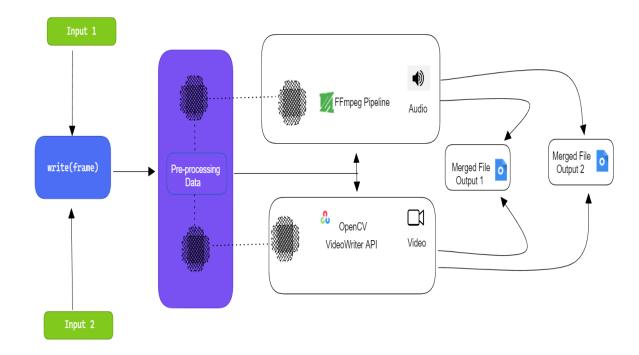
Architecture and Design

- Client devices: These are the devices that connect to the streaming server to receive the video feed. They could be smartphones, laptops, or other internet-enabled devices.
- **Streaming server**: The server that sends the video stream to the clients. It receives the video input from a camera or other sources, processes it, and sends it out to the clients.
- OpenCV and FFmpeg: These are libraries used for image and audio processing, respectively. They are used to capture and process the video stream from the source.

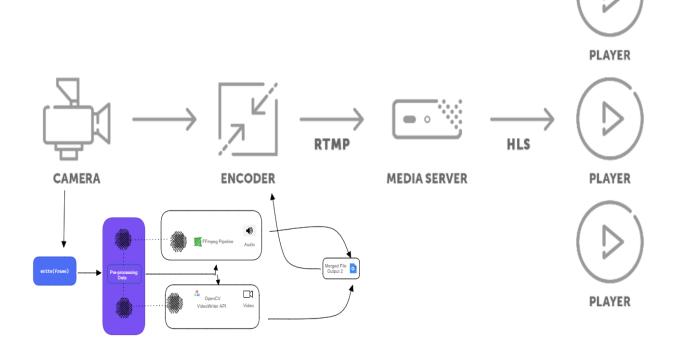
Architecture and Design

- Flask: Flask is a web framework used to develop server-side web applications in Python. It have been used to integrate OpenCV and FFmpeg with the streaming server.
- RTMP and HLS: These are protocols used for video streaming. RTMP
 (Real-Time Messaging Protocol) is used for low-latency streaming, while
 HLS (HTTP Live Streaming) is used for adaptive bitrate streaming.
- **Nginx server**: Nginx is a popular web server used to host websites and applications. It has been used to support for RTMP/HLS streaming.

Flowchart Of OpenCV & FFmpeg



RTMP && HLS WorkFlow



Challenges

- **1. Bandwidth management**: Streaming video to multiple clients simultaneously requires a lot of bandwidth, and managing this bandwidth can be a challenge.
- **2. Synchronization**: Ensuring that all clients receive the same stream at the same time is critical for a seamless experience. Any delay or lag in the stream can cause synchronization issues.
- **3. Encoding and decoding**: Encoding and decoding video and audio in real-time for multiple clients can be resource-intensive, requiring specialized hardware or software to handle the load.
- **4. Scalability:** As the number of clients increases, the system must be able to scale to handle the increased load without sacrificing performance.

Best practices:

- 1. Use adaptive bitrate streaming to ensure that clients receive the highest quality stream possible based on their available bandwidth.
- 2. Implement caching and load balancing to improve performance and reduce bandwidth usage.
- 3. Regularly monitor server performance, bandwidth usage, and client connection quality to identify and address potential issues.
- 4. Implement failover mechanisms to ensure uninterrupted streaming in the event of server or network failures.

The proposed approach is tested using Python programming language and a per- sonal computer with an Ryzen 7 4800H CPU @ 2.9 GHz processor and a 16 GB RAM. The operating system is Windows 11. The average latency time for multi- client streaming was 31 seconds.

 In this section, we will discuss results and performance of our technique and demonstrate the working.

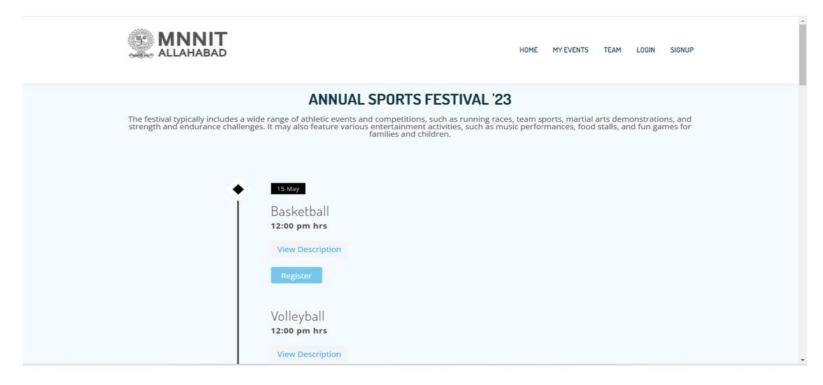


Fig 1: Home Page

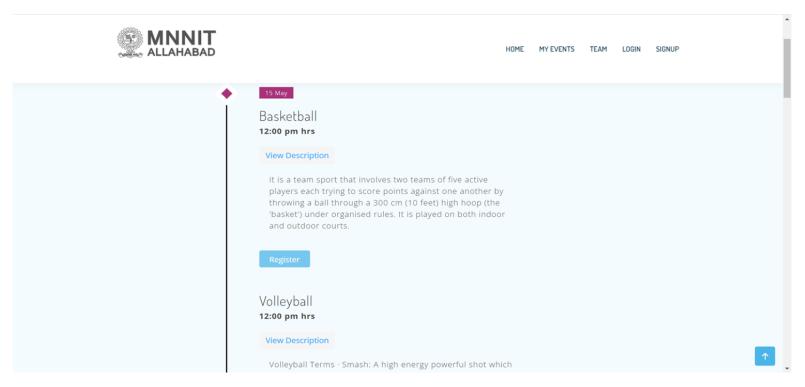


Fig 2: Home Page

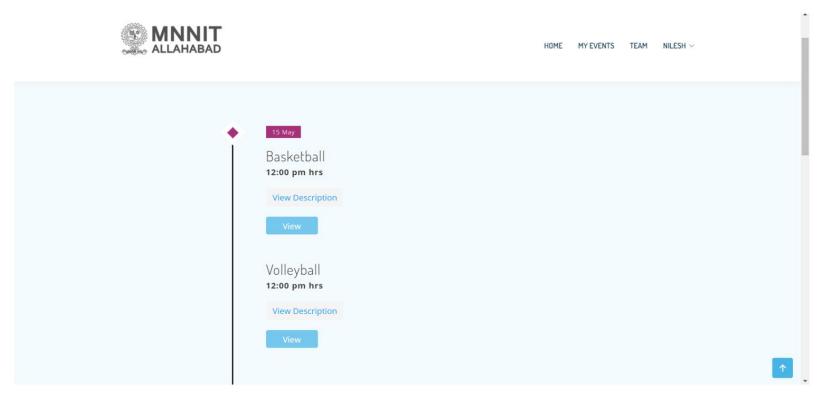


Fig 3: User Events Page

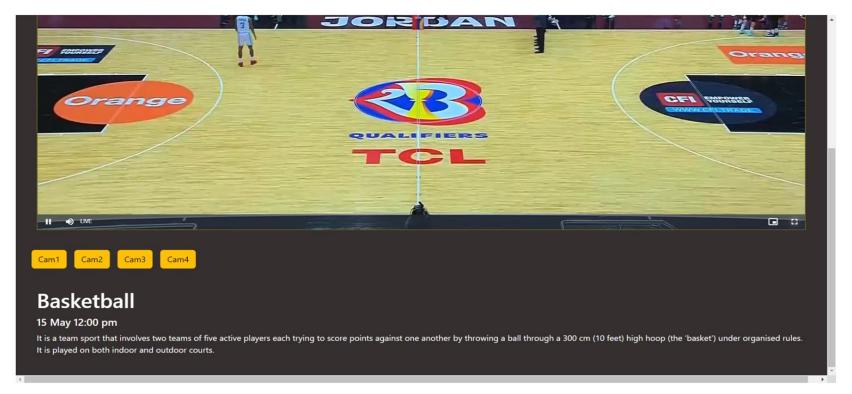


Fig4: Live Stream Page

- Minimum Latency: Measure the latency of the streaming platform by monitoring the time it takes for a video frame to travel from the source to the client devices. Use tools or methods to measure the round-trip time (RTT) or time taken for a packet to travel from the source to the client and back. Assess the platform's ability to minimize latency by optimizing the streaming pipeline, network configurations, and using low-latency streaming protocols like WebRTC or MPEG-DASH with low-latency modes.
- Bandwidth: Conduct bandwidth testing to determine the maximum throughput of the
 network connection. Use network testing tools or services to measure the upload and
 down- load speeds available for streaming. Test the streaming platform's ability to adapt to
 different bandwidth conditions by simulating low-bandwidth or high-bandwidth scenarios
 and evaluating the platform's performance in delivering video streams with appropriate
 quality levels.

 Buffer Size: Test the buffer handling under different net- work conditions, such as varying bandwidth or intermittent connectivity.

Parameter	OpenCV	FFmpeg	OpenMediaStream
Min. Latency	25s	38s	78s
BandWidth	700 - 1000 KBPS	400-500 KKBPS	500 - 800 KBPS
Buffer size	600 - 1000 frame		

Table1: Parametric Analysis on various technologies

Future Direction

- Reducing latency in multiclient streaming to enhance the real-time experience for viewers by using techniques like Adaptive Bitrate Streaming, Low-Latency Streaming Protocols and Caching and Prefetching.
- Enhancing video quality by Video Transcoding and Preprocessing.

Conclusion

Efficient Video Streaming: By leveraging Nginx as a web server and RTMP
as a streaming protocol, we have established a robust and scalable
infrastructure for delivering video content to multiple clients simultaneously.

 Flexibility and Compatibility: With the support of FFmpeg, we have achieved broad compatibility with various video formats and codecs.

 Adaptive Bitrate Streaming: The implementation of HLS (HTTP Live Stream- ing) has enabled adaptive bitrate streaming, allowing clients to dynamically adjust the video quality based on their network conditions.

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Thank You!