VIETNAM NATIONAL UNIVERSITY, HO CHI MINH CITY UNIVERSITY OF TECHNOLOGY FACULTY OF COMPUTER SCIENCE AND ENGINEERING



Computer Network (CO3094)

Assignment 1

Real-Time Streaming Protocol (RTSP) and Real-time Transfer Protocol (RTP)

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	the user. Given that SETUP is mandatory in an RTSP interaction, how would you implement that in a media player? When does the client send the SETUP? Come up with a solution and implement it. Also, is it appropriate to send TEARDOWN when the user clicks on the STOP button?	9				
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1 Software Requirements Analysis

1.1 Functional Requirements

1.1.0.a System Requirements

- The system can work (stream video).
- The system can communicate with users via RTSP/RTP protocol.

1.1.0.b Client Requirements

- It is possible to connect to the server via the terminal.
- User can play video from server, stop and end.
- User can view basic video parameters such as video duration.

1.2 Non-functional Requirements

- Videos must be in .Mjpeg format
- Response time from server ≤ 0.5 s



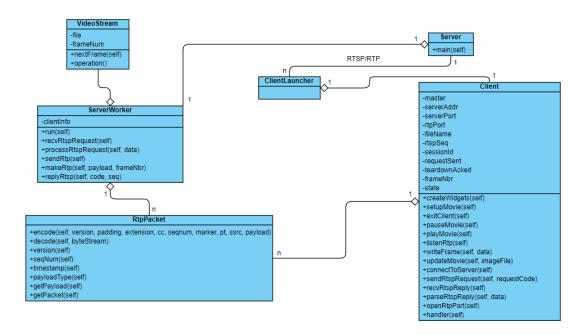
2 Description of different functions for your application

Class Name	Function	Parameter	Description
	init(self, master,	self, master,	
	serveraddr, serverport,	serveraddr, serverport,	Constructor
Client	rtpport, filename)	rtpport, filename	
	createWidgets(self)	self	Create GUI
	setupMovie(self)	self	Setup button handler
	exitClient(self)	self	Teardown button handler
	pauseMovie(self)	self	Pause button handler
	playMovie(self)	self	Play button handler
	listenRtp(self)	self	Listen for RTP packets
	writeFrame(self, data)	self, data	Write the received frame to a temp image file. Return the image file.
	updateMovie(self, imageFile)	self, imageFile	Update the image file as video frame in the GUI
	connectToServer(self)	self	Connect to the Server, Start a new RTSP/TCP session.
	sendRtspRequest(self, requestCode)	self, requestCode	Send RTSP request to the server.
	recvRtspReply(self)	self	Receive RTSP reply from the server.
	parseRtspReply(self, data)	self, data	Parse the RTSP reply from the server.
	openRtpPort(self)	self	Open RTP socket binded to a specified port.
	handler(self)	self	Handler on explicitly closing the GUI window.
ClientLauncher	main()		testoase
ServerWorker	init(self, clientInfo)	self, clientInfo	Constructor
	run(self)	self	Run the server
	recvRtspRequest(self)	self	Receive RTSP request from the client.
	processRtspRequest(self, data)	self, data	Process RTSP request sent from the alient.
	sendRtp(self)	self	Send RTP packets over UDP:
	makeRtp(self, payload, frameNbr)	self, payload, frameNbr	RTP-packetize the video data
	replyRtsp(self, code, seq)	self, code, seq	Send RTSP reply to the client.
Server	main(self)	self	Main function to run the whole program
RtpPacket	init(self)	self	Constructor
	encode(self, version, padding,	self, version, padding,	
	extension, co, segnum, marker, pt, ssrc,		Encode the RTP packet with header fields and pauload
	payload)	pt, ssrc, payload	Litode the ritr packet likirineaderheids and payload
	1 1		
	decode(self, byteStream)	self, byteStream	Decode the RTP packet.
	version(self)	self	Return RTP version.
	seqNum(self)	self	Return sequence (frame) number.
	timestamp(self)	self	Return timestamp
	payloadType(self)	self	Return payload type.
	getPayload(self)	self	Return payload.
	getPacket(self)	self	Return RTP packet.
VideoStream	init(self, filename)	self, filename	Constructor
	nextFrame(self)	self	4
	frameNbr(self)	self	Get frame number.

Figure 1 – Function Description



3 Class diagram



 $Figure \ 2-Class \ Diagram$



4 A summative evaluation of achieved results

- Complete the RTSP protocol in the client
- Complete the RTP packetization in the server (named as RtpPacket.py)
- Make the customized interactive player for displaying the transmitted video



5 User manual

— Step 1: First, we run terminal and start the server with the command: **python Server.py server port**.

We should set the port number larger than 1024.

Example:



FIGURE 3 – Run server with port = 1300

— Step 2, start the client with the command **python ClientLauncher.py server-host server-port RTP-port video-file** where server host is the name of the machine where the server is running, server port is the port where the server is listening on, RTP port is the port where the RTP packets are received, and video file is the name of the video file you want to request (we have provided one example file movie.Mjpeg).

Example:

— server-host: LAPTOP-MMCHUE76 (use cmd to check)

— sever-port : 1300 (create in first step)

— RTP-port : We choose 6000— video-file : movie.Mjpeg

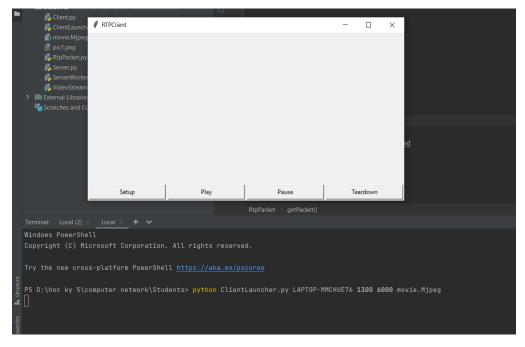


FIGURE 4 - Run Client



— Step 3, click **Setup** to create RTP connection.

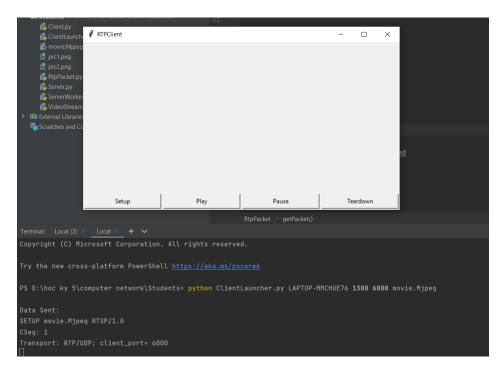


Figure 5 – Setup

— Final, we can press **Play** to play video, **Pause** to stop and **Teardown** to close the client UI.

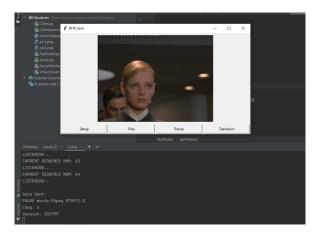


Figure 6 – Pause

— New GUI:



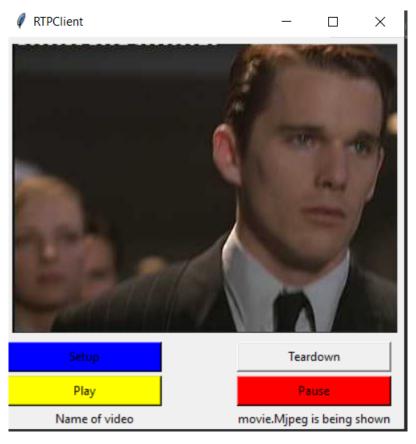


Figure 7 - new GUI



6 Extend

6.1 Calculate the statistics about the session. You will need to calculate RTP packet loss rate, video data rate(in bits or bytes per second), and any other interesting statistics that you can think of.

We have add some calculation for the code to calculate some statistics, which display in terminal after we guit the application.(data rate, packetlost, time)



Figure 8 – statistics

6.2 The user interface on the RTPClient has 4 buttons for the 4 actions. If you compare this to a standard media player, such as RealPlayer or Windows Media Player, you can see that they have only 3 buttons for the same actions: PLAY, PAUSE, and STOP (roughly corresponding to TEARDOWN). There is no SETUP button available to the user. Given that SETUP is mandatory in an RTSP interaction, how would you implement that in a media player? When does the client send the SETUP? Come up with a solution and implement it. Also, is it appropriate to send TEARDOWN when the user clicks on the STOP button?

i would implement a state machine with 3 state : init, playing and paused as below :

The client will send SETUP right after we pressed play button. It is also appropriate to send

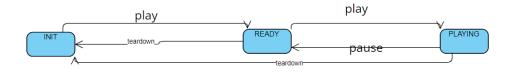


Figure 9 – FSM

TEARDOWN when user clicks on STOP button.



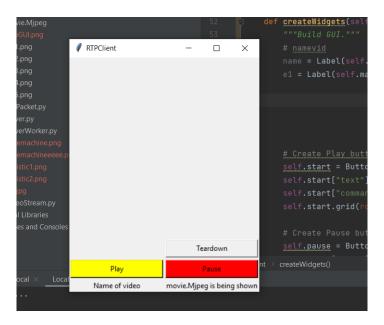


Figure 10 - scrs1

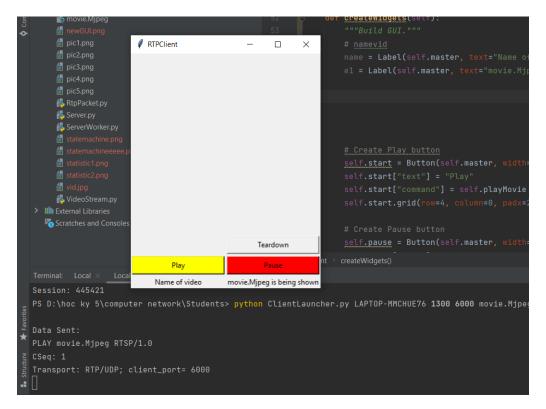


Figure 11 – scrs2



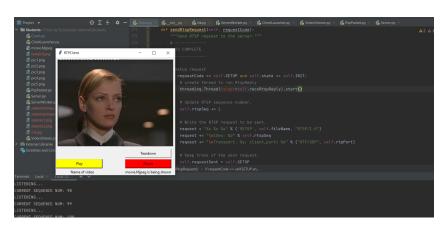


Figure 12 – scrs3



6.3 Currently, the client and server only implement the minimum necessary RTSP interactions and PAUSE. Implement the method DESCRIBE which is used to pass information about the media stream. When the server receives a DESCRIBE-request, it sends back a session description file which tells the client what kinds of streams are in the session and what encodings are used.

When pressed Describe button, the information of video is displayed at right corner.

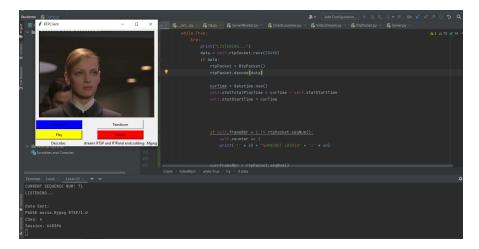


Figure 13 - scrs4



- 6.4 Implement some additional functions for user interface such as: display video total time and remaining time, fast forward or backward video (or make a scroll bar for scrolling video if you can)
- 6.5 Add one more state to the client (for example SWITCH state) so that user can select another video from a list of videos received from server.



7 Full source code

Github: https://github.com/minhle270901/1952844.git