

Video Server and Client

Submitted to:

Prof. Cherine Saleh Eng. Marwan Nour

Submitted by:

Shady Tarek 19100178
Ahmed Khairy 19100561
Kareem Akram 19102827
Yousef Shalaby 19103854
Raneem Abdelwahab 19101342
Aya Barakat 19102308

Project description

The project talks about a video library stored in a server that can be accessed by clients where he sends a request on TCP module then receives the video and get watched on the client device.

A video streaming server is an application that allows users to access and stream video content from a central location. This can be achieved by having one computer act as a server that stores a library of videos. Other computers can then connect to this server and stream the videos in real-time.

The server can be set up using various programming languages and technologies. For example, it can be created using Python and socket programming or by leveraging existing libraries such as (python-ffmpeg-video-streaming) and (pylivestream). These libraries provide tools and functionalities to help package media content for online streaming and stream to one or multiple sites simultaneously.

Once the server is set up, users can connect to it using a client application. This client application can be developed for various platforms such as web browsers, mobile devices, and smart TVs. The client application communicates with the server to request and receive video streams.

The video streaming server can also implement features such as user authentication, video search, and video recommendations. This allows for a personalized and user-friendly experience like popular video streaming platforms like YouTube.

Snips from server code (local host):

```
import socket,cv2, pickle,struct
import pyshine as ps # pip install pyshine
import cv2, import import numpy as np
import numpy as np
import threading, wave, pyaudio,pickle,struct
import sys
import queue
import os
from moviepy.editor import VideoFileClip
from os.path import exists

client_socket = socket.socket(socket.AF_INET,socket.SOCK_STREAM)
host_ip = 'localhost' # Here according to your server ip write the address

port = 18000
client_socket.connect((host_ip,port))

def convert_video_to_audio_moviepy(video_file, output_ext="wav"):
    """Converts video to audio using MoviePy library
    that uses 'ffmpeg' under the hood""
    flename, ext = os.path.splitext(video_file)
    clip = VideoFileClip(video file)
    clip = uidoFileClip(video file)
    clip.audio.write_audiofile(f"{filename}.{output_ext}")

isExistingVid1 = os.path.exists('Videos\bmwd.mp4')
isExistingVid2 = os.path.exists('Videos\bmwd.mp4')
isExistingVid3 = os.path.exists('Videos\bmwd.mp4')
isExistingVid4 = os.path.exists('Videos\bmwd.mp4')
```

```
import socket,cv2, pickle,struct
import pyshine as ps # pip install pyshine
import imutils # pip install imutils
import time
import cv2, imutils, socket
import numpy as np
import base64
import threading, wave, pyaudio, pickle, struct
import queue
import os
from moviepy.editor import VideoFileClip
from os.path import exists
client_socket = socket.socket(socket.AF_INET,socket.SOCK_STREAM)
host_ip = 'localhost' # Here according to your server ip write the address
port = 18000
client_socket.connect((host_ip,port))
def convert_video_to_audio_moviepy(video_file, output_ext="wav"):
     """Converts video to audio using MoviePy library
    that uses `ffmpeg` under the hood""
    filename, ext = os.path.splitext(video_file)
    clip = VideoFileClip(video_file)
    clip.audio.write_audiofile(f"{filename}.{output_ext}")
isExistingVid1 = os.path.exists('Videos/bmwd.mp4')
isExistingVid2 = os.path.exists('Videos\Range Rover Velar.mp4')
isExistingVid3 = os.path.exists('Videos\mer1.mp4')
isExistingVid4 = os.path.exists('Videos\Porsche.mp4')
```

```
if isExistingVid1==False:
   convert_video_to_audio_moviepy('Videos/bmwd.mp4')
if isExistingVid2==False:
    convert_video_to_audio_moviepy('Videos\Range Rover Velar.mp4')
if isExistingVid3==False:
   convert_video_to_audio_moviepy('Videos\mer1.mp4')
if isExistingVid4==False:
   convert_video_to_audio_moviepy('Videos\Porsche.mp4')
def video():
   while True:
    if client_socket:
       msg=client_socket.recv(5)
        print("\n","VIDEO",msg)
        if msg==b'v1':
           vid = cv2.VideoCapture('Videos/bmwd.mp4')
        elif msg==b'v2':
           vid = cv2.VideoCapture('Videos\Range Rover Velar.mp4')
        elif msg==b'v3':
            vid = cv2.VideoCapture('Videos\mer1.mp4')
        elif msg==b'v4':
            vid = cv2.VideoCapture('Videos\Porsche.mp4')
        elif msg==b'bye':
           client_socket.close()
           continue
```

```
while (vid.isOpened()):
             try:
                 img, frame = vid.read()
frame = imutils.resize(frame,width=380)
                 a = pickle.dumps(frame)
                 message = struct.pack("Q",len(a))+a
                 time.sleep(0.031)
                 client_socket.sendall(message)
                 key = cv2.waitKey(1) & 0xFF
                 if key == ord("q"):
                     client_socket.close()
             except:
                 print('VIDEO FINISHED!')
clientAudio_socket = socket.socket(socket.AF_INET,socket.SOCK_STREAM)
clientAudio_socket.connect((host_ip,port-1))
def audio():
    CHUNK = 1024
    p = pyaudio.PyAudio()
    while True:
        if clientAudio_socket:
            msg=clientAudio_socket.recv(10)
             print("\n","AUDIO",msg)
if msg==b'v1':
```

```
def audio():
    p = pyaudio.PyAudio()
         if clientAudio_socket:
              msg=clientAudio_socket.recv(10)
              print("\n","AUDIO",msg)
if msg==b'v1':
                  wf = wave.open("Videos/bmwd.wav", 'rb')
              elif msg==b'v2':
                  wf = wave.open("Videos\Range Rover Velar.wav", 'rb')
              elif msg==b'v3':
                  wf = wave.open("Videos/mer1.wav", 'rb')
              elif msg==b'v4':
                  wf = wave.open("Videos/Porsche.wav", 'rb')
              elif msg==b'bye':
                  clientAudio_socket.close()
              stream = p.open(format=p.get_format_from_width(wf.getsampwidth()),
                       channels=wf.getnchannels(),
rate=wf.getframerate(),
                       input=True,
frames_per_buffer=CHUNK)
                  data = wf.readframes(CHUNK)
                  a = pickle.dumps(data)
message = struct.pack("Q",len(a))+a
clientAudio_socket.sendall(message)
```

Snips from server code (Eathernet):

```
client_socket = socket.socket(socket.AF_INET,socket.SOCK_STREAM)
host_ip = '192.168.33.100' # Here according to your server ip write the address

port = 18000
client_socket.connect((host_ip,port))

def convert_video_to_audio_moviepy(video_file, output_ext="wav"):
    """Converts video to audio using MoviePy library
    that uses `ffmpeg` under the hood"""
    filename, ext = os.path.splitext(video_file)
    clip = VideoFileClip(video_file)
    clip.audio.write_audiofile(f"{filename}.{output_ext}")

isExistingVid1 = os.path.exists('Videos/bmwd.mp4')
isExistingVid2 = os.path.exists('Videos/Range Rover Velar.mp4')
isExistingVid3 = os.path.exists('Videos\Porsche.mp4')

if isExistingVid4 = os.path.exists('Videos\Porsche.mp4')

if isExistingVid2==False:
    convert_video_to_audio_moviepy('Videos\Porsche.mp4')

if isExistingVid3==False:
    convert_video_to_audio_moviepy('Videos\Porsche.mp4')

if isExistingVid3==False:
    convert_video_to_audio_moviepy('Videos\Porsche.mp4')

if isExistingVid4==False:
    convert_video_to_audio_moviepy('Videos\Porsche.mp4')

current_proc = None
```

```
def stopCurrentVid():
    global current proc
    if current_proc and current_proc.poll() is None:
        current proc.kill()
def playVideo(vid):
   global current proc
    if playing_video:
        stopCurrentVid()
    playing video = True
    while (vid.isOpened()):
            try:
                img, frame = vid.read()
                frame = imutils.resize(frame, width=380)
                a = pickle.dumps(frame)
                message = struct.pack("Q",len(a))+a
                time.sleep(0.031)
                client socket.sendall(message)
                key = cv2.waitKey(1) & 0xFF
                if key == ord("q"):
                    client socket.close()
            except:
                print('VIDEO FINISHED!')
                break
```

```
def video():
     if client_socket:
        msg=client_socket.recv(5)
        print("\n","VIDEO",msg)
        if msg==b'v1':
           vid = cv2.VideoCapture('Videos/bmwd.mp4')
           playVideo(vid)
        elif msg==b'v2'
           vid = cv2.VideoCapture('Videos\Range Rover Velar.mp4')
           playVideo(vid)
        elif msg==b'v3':
           vid = cv2.VideoCapture('Videos\mer1.mp4')
           playVideo(vid)
        elif msg==b'v4':
           vid = cv2.VideoCapture('Videos\Porsche.mp4')
           playVideo(vid)
clientAudio_socket = socket.socket(socket.AF_INET,socket.SOCK_STREAM)
clientAudio_socket.connect((host_ip,port-1))
```

```
def playaudio(wf):
   CHUNK = 1024
   p = pyaudio.PyAudio()
   stream = p.open(format=p.get format from width(wf.getsampwidth()),
                    channels=wf.getnchannels(),
                    rate=wf.getframerate(),
                    input=True,
                    frames per buffer=CHUNK)
   while True:
     try:
       data = wf.readframes(CHUNK)
        a = pickle.dumps(data)
       message = struct.pack("Q",len(a))+a
       clientAudio socket.sendall(message)
      except:
                print('Audio FINISHED!')
                break
```

```
def audio():
        if clientAudio socket:
            msg=clientAudio_socket.recv(10)
            print("\n","AUDIO",msg)
if msg==b'v1':
                wf = wave.open("Videos/bmwd.wav", 'rb')
                t1 = threading.Thread(target=playaudio(wf))
                t1.start()
            elif msg==b'v2':
                wf = wave.open("Videos\Range Rover Velar.wav", 'rb')
                t1 = threading.Thread(target=playaudio(wf) )
                t1.start()
            elif msg==b'v3':
               wf = wave.open("Videos/mer1.wav", 'rb')
                playaudio(wf)
            elif msg==b'v4':
                wf = wave.open("Videos/Porsche.wav", 'rb')
            playaudio(wf)
elif msg==b'bye':
                clientAudio_socket.close()
            break
```

```
from concurrent.futures import ThreadPoolExecutor
with ThreadPoolExecutor(max_workers=3) as executor:
    executor.submit(audio)
    executor.submit(video)
```

Client code:

```
Gulpy > 1:10 > Octobridgets import *

from PyQt5.Qttdigets import *

from PyQt5.Qttdi import *

from PyQt5.Qttdi import *

from PyQt5.Qttdi import *

from PyQt5.Stuport Qttore, Qttdigets,Qtdi , QtMultimedia

import sys

import subprocess

from threading import *

from multiprocessing.connection import Listener

from widgear.gears import MetGear

from imutils import build_montages

import cv2

import cv2

import pygame

from socket import socket

import socket

import socket

import ptckle

import time

import time

import time

import true,

import time

import true,

import threading, wave, pyaudio,pickle,struct, queue

from PyQt5.QtCore import datetime

import to

import to

import to

import to

import o

import designer

impor
```

```
def clickvideo1(self):
    if self.video=True:
        # if self.client_socket and self.clientAudio_socket:
        # self.client_socket.send(bytes('bye',"utf-8"))
        # self.clientAudio_socket.send(bytes('bye',"utf-8"))
        self.clientAudio_socket.close()
        self.clientAudio_socket.close()
        # else:
        # self.clientAudio_socket.send(bytes('v1',"utf-8"))
        # self.clientAudio_socket.send(bytes('v1',"utf-8"))
        cmd='python client_1.py'
        self.client_l=subprocess.Popen(cmd,shell=True)
    self.video=Irue:
        # if self.client_l=subprocess.Popen(cmd,shell=True)

def clickvideo2(self):
        if self.client_socket and self.clientAudio_socket:
        # self.client_socket.send(bytes('bye',"utf-8"))
        # self.client_socket.send(bytes('bye',"utf-8"))
        self.client_socket.close()
        self.client_socket.close()
        # else:
        # self.clientAudio_socket.close()
        # self.client_socket.send(bytes('v2',"utf-8"))
        # self.clientAudio_socket.send(bytes('v2',"utf-8"))
        cmd='python client_l.py'
        self.client_socket.send(bytes('v2',"utf-8"))
        cmd='python client_l.py'
        self.c
```

```
def clickvideo3(self):
    if self.video==True:

    # self.client_socket.send(bytes('bye',"utf-8"))
    # self.clientAudio_socket.send(bytes('bye',"utf-8"))
    self.clientAudio_socket.close()
    self.client_socket.close()
    self.client_1=subprocess.Popen(cmd,shell=True)
    self.v='v3'
    # cmd='python client_3.py'
    # self.client_3=subprocess.Popen(cmd,shell=True)

def clickvideo4(self):
    if self.video==True:

    # self.client_socket.send(bytes('bye',"utf-8"))
    # self.clientAudio_socket.send(bytes('bye',"utf-8"))
    self.clientAudio_socket.close()
    cmd='python client_1.py'
    self.client_1=subprocess.Popen(cmd,shell=True)
    self.client_1=subprocess.Popen(cmd,shell=True)
    self.v='v4'

def AudioClient_thread(self):
    # server_socket = socket.socket(socket.AF_INET,socket.SOCK_STREAM)
    # host_name = socket.gethostname()
    host_ip = 'localhost'#socket.gethostbyname(host_name) 192.168.33.100
    # print('NOST IP:', host_ip)
    port = 18000
    # socket_address = (host_ip,port)
    # server_socket.bind(socket_address)
    # server_socket.listen()
    # print("Listening at",socket_address)
    # create socket
    serverAudio_socket = socket.socket(socket.AF_INET,socket.SOCK_STREAM)
    socketAudio_address = (host_ip,port-1)
```

```
# Create socket
serverAudio_socket = socket.socket(socket.AF_INET_socket.SOCK_STREAM)
socketAudio_address = (host_ip,port-1)
print('server_listening at', socketAudio_address)
# serverAudio_socket.commect(socketAudio_address)
# serverAudio_socket.bind(socketAudio_address)
serverAudio_socket.listen()
print("CLIENT_COMMECTED_TO", socketAudio_address)

while True:
# self.client_socket,addr = server_socket.accept()
self.clientAudio_socket,addr2 = serverAudio_socket.accept()
# print(self.clientAudio_socket)

# thread = threading.Thread(target=self.show_client, args=(addr,self.client_socket))
# thread_start()

thread2 = threading.Thread(target=self.audio_stream, args=(addr_self.clientAudio_socket))
thread2.start()

print("TOTAL_Audio_CLIENTS ", threading.active_count() - 1)

def Client_thread(self):
server_socket = socket.socket(socket.AF_INET_socket.SOCK_STREAM)
host_name = socket.gethostname()
host_ip = 'localhost'=socket.gethostbyname(host_name) 192.168.33.100
print('HOST_IP:',host_ip)
port = 18000
socket_address = (host_ip,port)
server_socket.bind(socket_address)
# create socket
# serverAudio_socket_address)
# create socket
# serverAudio_socket_address)
# create socket
# serverAudio_socket_connect(socketAAF_INET_socket.SOCK_STREAM)
# socketAudio_sodress = (host_ip,port-1)
# print("Listening_at", socketAudio_address)
# serverAudio_socket_connect(socketAudio_address)
# serverAudio_socket.connect(socketAudio_address)
```

```
def audio_stream(self,addr2,clientAudio_socket):
   stream = p.open(format=p.get_format_from_width(2),
                   channels=2,
                   rate=44100.
                   frames_per_buffer=CHUNK)
   payload_size = struct.calcsize("Q")
   while True:
           if self.clientAudio_socket :
               self.clientAudio_socket.send(bytes(self.v,"utf-8"))
               self.clientAudio_socket.send(bytes(self.v,"utf-8"))
               data+=packet
           packed_msg_size = data[:payload_size]
           data = data[payload_size:]
           msg_size = struct.unpack("Q",packed_msg_size)[0]
           while len(data) < msg_size:
              data += self.clientAudio_socket.recv(100)
           frame_data = data[:msg_size]
           data = data[msg size:]
           frame = pickle.loads(frame data)
           stream.write(frame)
```

```
| Scut, | Scut
```

```
data+=packet
packed_msg_size = data[:payload_size]
data = data[payload_size:
msg_size = struct.unpack("Q",packed_msg_size)[0]
while len(data) < msg_size:
    data += self.client_socket.recv(2000000)
frame_data = data[:msg_size]
data = data[msg_size:]
frame = pickle.loads(frame_data)
text = f"CLIENT: {addr}"
# time_now = datetime.now().strftime("%d/%m/%y %H:%M:%S")
# frame = ps.putBText(frame,time_now,10,10,vspace=10,hspace=1,font_scale=0.7, background_RGB=(141,0,0)
if not frame_shape:
    # video_file_name = str(addr) + time_name
# out = cv2.VideoWriter(video_file_name, fourcc, fps, (frame.shape[1], frame.shape[0]), True)
    frame_shape = True
# out.write(frame)
# cv2.imshow(f"FROM {addr}",frame)
frame= cv2.resize(frame,[765,575])
self.Camera_representation.setPixmap(QPixmap.fromImage(converted))
```

Summary:

In summary, a video streaming server is an application that allows users to access and stream video content from a central location. It can be set up using various programming languages and technologies and provides a personalized and user-friendly experience for its users.