FFmpeg开发常用功能封装

2023-08-24 16:33 · 音视频开发老舅

 \star

开发中使用到的FFmpeg常用功能,避免相同功能代码的重复编写,使用时直接复制 提升效率。由于音视频处理的场景众多,无法编写完全通用的方法接口,可能需根据 实际场景进行一定的修改,本文章中的代码也将持续更新优化。

这里提供ffmpegheader.h,ffmpegheader.cpp。配置好基本的FFmpeg库环境后, 直接导入上述两个文件,即可直接使用对应功能。

▶ 1、头文件

```
#ifndef FFMPEGHEADER_H
#define FFMPEGHEADER_H
* 封装常用的ffmpeg方法以及类 只需要引入文件即可直接使用 避免重复轮子
* By ZXT
extern "C"
#include "./libavcodec/avcodec.h"
#include "./libavformat/avformat.h"
#include "./libavformat/avio.h"
#include "./libavutil/opt.h"
#include "./libavutil/time.h"
#include "./libavutil/imqutils.h"
#include "./libswscale/swscale.h"
#include "./libswresample/swresample.h"
#include "./libavutil/avutil.h"
#include "./libayutil/ffversion.h"
#include "./libavutil/frame.h"
#include "./libavutil/pixdesc.h"
#include "./libavutil/imgutils.h"
#include "./libavfilter/avfilter.h"
#include "./libavfilter/buffersink.h"
#include "./libavfilter/buffersrc.h"
#include "./libavdevice/avdevice.h"
#include <QDebug>
```

/****** 常用函数封装 *************

//获取ffmpeg报错信息

char *getAVError(int errNum);

//根据pts计算实际时间us

int64_t getRealTimeByPTS(int64_t pts, AVRational timebase);

//pts转换为us差时进行延时

void calcAndDelay(int64_t startTime, int64_t pts, AVRational timebase);

//十六进制字节数组转十进制

int32_t hexArrayToDec(char *array, int len);

/*******

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//视频图像转换类

class VideoSwser public: VideoSwser(); ~VideoSwser();

//初始化转换器

bool initSwsCtx(int srcWidth, int srcHeight, AVPixelFormat srcFmt, int dstWidth, in void release();

//返回转换格式后的AVFrame

AVFrame *getSwsFrame(AVFrame *srcFrame); private:

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```
bool hasInit;
 bool needSws;
 int dstWidth;
 int dstHeight;
 AVPixelFormat dstFmt;
//格式转换
 SwsContext *videoSwsCtx;
 };
//视频编码器类
 class VideoEncoder
 {
 public:
 VideoEncoder();
 ~VideoEncoder();
//初始化编码器
 bool initEncoder(int width, int height, AVPixelFormat fmt, int fps);
 void release();
//返回编码后AVpacket
 AVPacket *getEncodePacket(AVFrame *srcFrame);
 AVPacket *flushEncoder();
//返回编码器上下文
 AVCodecContext *getCodecContent();
 private:
 bool hasInit;
//编码器
 AVCodecContext *videoEnCodecCtx;
};
//音频重采样类
 class AudioSwrer
 public:
 AudioSwrer();
 ~AudioSwrer();
//初始化转换器
 bool initSwrCtx(int inChannels, int inSampleRate, AVSampleFormat inFmt, int outChar
 void release();
//返回转换格式后的AVFrame
 AVFrame *getSwrFrame(AVFrame *srcFrame);
//返回转换格式后的AVFrame srcdata为一帧源格式的数据
 AVFrame *getSwrFrame(uint8_t *srcData);
 private:
 bool hasInit;
 bool needSwr;
 int outChannels;
 int outSampleRate;
 AVSampleFormat outFmt;
//格式转换
 SwrContext *audioSwrCtx;
 };
//音频编码器类
 class AudioEncoder
 ł
 public:
 AudioEncoder();
 ~AudioEncoder();
//初始化编码器
 bool initEncoder(int channels, int sampleRate, AVSampleFormat sampleFmt);
```

void release();

//返回编码后AVpacket

```
AVPacket *getEncodePacket(AVFrame *srcFrame);
AVPacket *flushEncoder();
```

//返回编码器上下文

```
AVCodecContext *getCodecContent();
private:
bool hasInit;
```

//编码器

```
AVCodecContext *audioEnCodecCtx;
};
```

//实时采集场景时间戳处理类

```
class AVTimeStamp
{
public:
```

//累加帧间隔 优点:时间戳稳定均匀 缺点:实际采集帧率可能不稳定,固定累加或忽略小数会累加误差造成不同步

//实时时间戳 优点:时间戳保持实时及正确 缺点:存在帧间隔不均匀,极端情况不能 正常播放

//累加+实时矫正 优点:时间戳实时且较为均匀 缺点:纠正时间戳的某一时刻可能画面或声音卡顿

```
erum PTSMode
{

PTS_RECTIFY = 0, //默认矫正类型 保持帧间隔尽量均匀

PTS_REALTIME //实时pts
};
public:
AVTimeStamp();
~AVTimeStamp();
```

//初始化时间戳参数

```
void initAudioTimeStampParm(int sampleRate, PTSMode mode = PTS_RECTIFY);
void initVideoTimeStampParm(int fps, PTSMode mode = PTS_RECTIFY);
```

//开始时间戳记录

```
void startTimeStamp();
```

//返回pts

```
int64_t getAudioPts();
int64_t getVideoPts();
private:
```

//当前模式

```
PTSMode aMode;
PTSMode vMode;
```

//时间戳相关记录 均us单位

```
int64_t startTime;
int64_t audioTimeStamp;
int64_t videoTimeStamp;
double audioDuration;
double videoDuration;
};
#endif // FFMPEGHEADER_H
```

▶ 2、实现文件

```
#include "ffmpegheader.h"
char *getAVError(int errNum)
{
    static char msg[32] = {0};
    ov_strerror(errNum, msg, 32);
    return msg;
}
int64_t getRealTimeByPTS(int64_t pts, AVRational timebase)
{
```

//pts转换为对应us值

```
AVRational timebase_q = {1, AV_TIME_BASE};
int64_t ptsTime = av_rescale_q(pts, timebase, timebase_q);
return ptsTime;
```

```
}
void calcAndDelay(int64_t startTime, int64_t pts, AVRational timebase)
{
int64_t ptsTime = getRealTimeByPTS(pts, timebase);
```

//计算startTime到此刻的时间差值

```
int64_t nowTime = av_gettime() - startTime;
int64_t offset = ptsTime - nowTime;
```

//大于2秒一般时间戳存在问题 延时无法挽救

```
if(offset > 1000 && offset < 2*1000*1000)
av_usleep(offset);
}
int32_t hexArrayToDec(char *array, int len)
{</pre>
```

//目前限制四字节长度 超过则注意返回类型 防止溢出

```
if(array == nullptr || len > 4)
int32_t result = 0;
for(int i=0; i<len; i++)</pre>
result = result * 256 + (unsigned char)array[i];
return result;
VideoSwser::VideoSwser()
videoSwsCtx = nullptr;
hasInit = false;
needSws = false;
VideoSwser::~VideoSwser()
release();
bool VideoSwser::initSwsCtx(int srcWidth, int srcHeight, AVPixelFormat srcFmt, int
release();
if(srcWidth == dstWidth \ \&\& \ srcHeight == dstHeight \ \&\& \ srcFmt == dstFmt)
needSws = false;
else
£
```

//设置转换上下文 srcFmt 到 dstFmt(一般为AV_PIX_FMT_YUV420P)的转换

```
videoSwsCtx = sws\_getContext(srcWidth, srcHeight, srcFmt, dstWidth, dstHeight, dstFreeDouble (srcWidth, dstFreeDouble (srcWidth) (srcWidth, dstFreeDouble (srcWidth) (srcWidth)
if (videoSwsCtx == NULL)
qDebug() << "sws_getContext error";</pre>
 return false;
this->dstFmt = dstFmt;
 this->dstWidth = dstWidth;
 this->dstHeight = dstHeight;
needSws = true;
hasInit = true;
 return true:
 void VideoSwser::release()
if(videoSwsCtx)
 sws_freeContext(videoSwsCtx);
videoSwsCtx = nullptr;
hasInit = false;
needSws = false;
 AVFrame *VideoSwser::getSwsFrame(AVFrame *srcFrame)
if(!hasInit)
qDebug() << "Swser未初始化";
 return nullptr;
if(!srcFrame)
 return nullptr;
if(!needSws)
 return srcFrame;
 AVFrame *frame = av_frame_alloc();
 frame->format = dstFmt;
 frame->width = dstWidth;
frame->height = dstHeight;
```

```
int ret = av_frame_get_buffer(frame, 0);
if (ret != 0)
{
    qDebug() << "av_frame_get_buffer swsFrame error";
    return nullptr;
}
ret = av_frame_make_writable(frame);
if (ret != 0)
{
    qDebug() << "av_frame_make_writable swsFrame error";
    return nullptr;
}
sws_scale(videoSwsCtx, (const uint8_t *const *)srcFrame->data, srcFrame->linesize,
    return frame;
}
VideoEncoder::VideoEncoder()
{
    videoEncodecCtx = nullptr;
    hasInit = false;
}
VideoEncoder::-VideoEncoder()
{
    release();
}
bool VideoEncoder::initEncoder(int width, int height, AVPixelFormat fmt, int fps)
{
```

//重置编码信息

```
release();
```

//设置编码器参数 默认AV_CODEC_ID_H264

```
AVCodec *videoEnCoder = avcodec_find_encoder(AV_CODEC_ID_H264);
if(!videoEnCoder)
{
    qDebug() << "avcodec_find_encoder AV_CODEC_ID_H264 error";
    return false;
}
videoEnCodecCtx = avcodec_alloc_context3(videoEnCoder);
if(!videoEnCodecCtx)
{
    qDebug() << "avcodec_alloc_context3 AV_CODEC_ID_H264 error";
    return false;
}
```

//重要! 编码参数设置 应根据实际场景修改以下参数

//设置QP最大和最小量化系数,取值范围为0~51 越大编码质量越差

```
videoEnCodecCtx->qmin = 10;
videoEnCodecCtx->qmax = 30;
```

//若设置此项 则sps、pps将保存在extradata; 否则放置于每个I帧前

```
videoEnCodecCtx->flags I= AV_CODEC_FLAG_GLOBAL_HEADER;
```

//预设参数 编码速度与压缩率的平衡 如编码快选择的算法就偏简单 压缩率低

//由慢到快veryslow slower slow medium fast faster veryfast superfast ultrafast 默认medium

```
int ret = av_opt_set(videoEnCodecCtx->priv_data, "preset", "ultrafast", 0);
if(ret != 0)
qDebug() << "av_opt_set preset error";</pre>
```

//偏好设置 进行视觉优化

//film电影 animation动画片 grain颗粒物 stillimage静止图片 psnr ssim图像评价指标 fastdecode快速解码 zerolatency零延迟

```
ret = av_opt_set(videoEnCodecCtx->priv_data, "tune", "zerolatency", 0);
if(ret != 0)
qDebug() << "av_opt_set preset error";
```

//画质设置 可能自动改变 如编码很快很难保证高画质会自动降级

```
//baseline实时通信 extended使用较少 main流媒体 high广电、存储
```

```
ret = av_opt_set(videoEnCodecCtx->priv_data, "profile", "main", 0);
if(ret != 0)
qDebug() << "av_opt_set preset error";</pre>
ret = avcodec_open2(videoEnCodecCtx, videoEnCoder, NULL);
if(ret != 0)
qDebug() << "avcodec_open2 video error";</pre>
return false;
hasInit = true;
return true;
void VideoEncoder::release()
if(videoEnCodecCtx)
avcodec_free_context(&videoEnCodecCtx);
videoEnCodecCtx = nullptr;
hasInit = false;
AVPacket *VideoEncoder::getEncodePacket(AVFrame *srcFrame)
if(!hasInit)
qDebug() << "VideoEncoder no init";</pre>
return nullptr;
if(!srcFrame)
return nullptr;
if(srcFrame->width != videoEnCodecCtx->width
| | srcFrame->height != videoEnCodecCtx->height
II srcFrame->format != videoEnCodecCtx->pix_fmt)
qDebug() << "srcFrame不符合视频编码器设置格式";
return nullptr:
```

//应保证srcFrame pts为us单位

```
srcFrame->pts = av_rescale_q(srcFrame->pts, AVRational{1, AV_TIME_BASE}, videoEnCod
int ret = avcodec_send_frame(videoEnCodecCtx, srcFrame);
if (ret != 0)
return nullptr;
```

//接收者负责释放packet

```
AVPacket *packet = av_packet_alloc();

ret = avcodec_receive_packet(videoEnCodecCtx, packet);

if (ret != 0)
{
    av_packet_free(&packet);
    return nullptr;
}

return packet;
}

AVPacket *VideoEncoder::flushEncoder()
{
    if(!hasInit)
{
    qDebug() << "VideoEncoder no init";
    return nullptr;
}

int ret = avcodec_send_frame(videoEnCodecCtx, NULL);

if (ret != 0)
    return nullptr;
```

//接收者负责释放packet

```
AVPacket *packet = av_packet_alloc();

ret = avcodec_receive_packet(videoEnCodecCtx, packet);

if (ret != 0) {

av_packet_free(&packet);

return nullptr;

}

return packet;

}

AVCodecContext *VideoEncoder::getCodecContent() {

return videoEnCodecCtx;

}

AudioSwrer::AudioSwrer() {

audioSwrCtx = nullptr;

hasInit = false;

needSwr = false;
```

```
AudioSwrer::~AudioSwrer()
release();
bool AudioSwrer::initSwrCtx(int inChannels, int inSampleRate, AVSampleFormat inFmt,
if(inChannels == outChannels && inSampleRate == outSampleRate && inFmt == outFmt)
needSwr = false;
else
audioSwrCtx = swr_alloc_set_opts(NULL, av_get_default_channel_layout(outChannels),
av_get_default_channel_layout(inChannels), inFmt, inSampleRate, 0, NULL);
if (!audioSwrCtx)
qDebug() << "swr_alloc_set_opts failed!";</pre>
return false;
int ret = swr_init(audioSwrCtx);
if (ret != 0)
qDebug() << "swr_init error";</pre>
swr_free(&audioSwrCtx);
return false;
this->outFmt = outFmt:
this->outChannels = outChannels:
this->outSampleRate = outSampleRate;
needSwr = true:
hasInit = true;
return true;
void AudioSwrer::release()
if(audioSwrCtx)
swr_free(&audioSwrCtx);
audioSwrCtx = nullptr;
hasInit = false;
needSwr = false:
AVFrame *AudioSwrer::getSwrFrame(AVFrame *srcFrame)
if(!hasInit)
qDebug() << "Swrer未初始化";
return nullptr;
if(!srcFrame)
return nullptr;
if(!needSwr)
return srcFrame;
AVFrame *frame = av_frame_alloc();
frame->format = outFmt;
frame->channels = outChannels;
frame->channel_layout = av_get_default_channel_layout(outChannels);
frame->nb_samples = 1024; //默认ad
int ret = av_frame_get_buffer(frame, 0);
if (ret != 0)
qDebug() << "av_frame_get_buffer audio error";</pre>
return nullptr;
ret = av_frame_make_writable(frame);
if (ret != 0)
qDebug() << "av_frame_make_writable swrFrame error";</pre>
return nullptr;
const uint8_t **inData = (const uint8_t **)srcFrame->data;
swr_convert(audioSwrCtx, frame->data, frame->nb_samples, inData, frame->nb_samples)
return frame:
AVFrame *AudioSwrer::getSwrFrame(uint8_t *srcData)
if(!hasInit)
qDebug() << "Swrer未初始化";
return nullptr;
if(!srcData)
return nullptr;
return nullptr;
AVFrame *frame = av_frame_alloc();
```

```
frame->channels = outChannels;
 frame->sample rate = outSampleRate:
 frame->channel_layout = av_get_default_channel_layout(outChannels);
 frame->nb_samples = 1024; //默认aad
 int ret = av_frame_get_buffer(frame, 0);
 if (ret != 0)
 qDebug() << "av_frame_get_buffer audio error";</pre>
 return nullptr;
 ret = av_frame_make_writable(frame);
 if (ret != 0)
 qDebug() << "av_frame_make_writable swrFrame error";</pre>
 return nullptr;
 const uint8_t *indata[AV_NUM_DATA_POINTERS] = {0};
 indata[0] = srcData;
 swr_convert(audioSwrCtx, frame->data, frame->nb_samples, indata, frame->nb_samples)
 AudioEncoder::AudioEncoder()
 audioEnCodecCtx = nullptr;
 hasInit = false;
 AudioEncoder::~AudioEncoder()
 bool AudioEncoder::initEncoder(int channels, int sampleRate, AVSampleFormat sampleF
4
```

//初始化音频编码器相关 默认AAC

//ffmpeg -h encoder=aac 自带编码器仅支持AV_SAMPLE_FMT_FLTP 大多数AAC 编码器都采用平面布局格式 提高数据访问效率和缓存命中率 加快编码效率

//音频数据量偏小 设置较为简单

```
audioEnCodecCtx->bit_rate = 64*1024;
audioEnCodecCtx->time_base = AVRational{1, sampleRate};
audioEnCodecCtx->sample_rate = sampleRate;
audioEnCodecCtx->sample_fmt = sampleFmt;
audioEnCodecCtx->channels = channels;
audioEnCodecCtx->channel_layout = av_get_default_channel_layout(channels);
audioEnCodecCtx->frame_size = 1024;
```

//打开音频编码器

```
int ret = avcodec_open2(audioEnCodecCtx, audioEnCoder, NULL);
if (ret != 0)
{
    qDebug() << "avcodec_open2 audio error" << getAVError(ret);
    return false;
}
hasInit = true;
return true;
}
void AudioEncoder::release()
{
    if(audioEnCodecCtx)
{
    avcodec_free_context(&audioEnCodecCtx);
    audioEnCodecCtx = nullptr;
}
hasInit = false;
}
AVPacket *AudioEncoder::getEncodePacket(AVFrame *srcFrame)
{
    if(!hasInit)
{
        qDebug() << "AudioEncoder no init";
        return nullptr;
}</pre>
```

```
}
if(!srcFrame)
return nullptr;
if(srcFrame->channels != audioEnCodecCtx->channels
!! srcFrame->sample_rate != audioEnCodecCtx->sample_rate
!! srcFrame->format != audioEnCodecCtx->sample_fmt)
{
qDebug() << "srcFrame-\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rightarrow\(\pi\)\rig
```

//应保证srcFrame pts为us单位

```
srcFrame->pts = av_rescale_q(srcFrame->pts, AVRational{1, AV_TIME_BASE}, audioEnCod
```

//进行音频编码得到编码数据AVPacket

```
int ret = avcodec_send_frame(audioEnCodecCtx, srcFrame);
if (ret != 0)
return nullptr;
```

//接收者负责释放packet

```
AVPacket *packet = av_packet_alloc();
ret = avcodec_receive_packet(audioEnCodecCtx, packet);
if (ret != 0)
{
    av_packet_free(&packet);
    return nullptr;
}
return packet;
}
AVPacket *AudioEncoder::flushEncoder()
{
    if(!hasInit)
{
        qDebug() << "AudioEncoder no init";
        return nullptr;
}
int ret = avcodec_send_frame(audioEnCodecCtx, NULL);
if (ret != 0)
return nullptr;
```

//接收者负责释放packet

```
AVPacket *packet = av_packet_alloc();

ret = avcodec_receive_packet(audioEnCodecCtx, packet);

if (ret != 0)
{
    av_packet_free(&packet);
    return nullptr;
}

return packet;
}

AVCodecContext *AudioEncoder::getCodecContent()
{
    return audioEnCodecCtx;
}

AVTimeStamp::AVTimeStamp()
{
    aMode = PTS_RECTIFY;
    vMode = PTS_RECTIFY;
    startTime = 0;
    audioTimeStamp = 0;
    videoTimeStamp = 0;
```

//默认视频264编码 25帧

```
videoDuration = 1000000 / 25;
```

//默认音频aac编码 44100采样率

```
audioDuration = 1000000 / (44100 / 1024);
}
AVTimeStamp::~AVTimeStamp()
{
}
void AVTimeStamp::initAudioTimeStampParm(int sampleRate, AVTimeStamp::PTSMode mode)
{
aMode = mode;
audioDuration = 1000000 / (sampleRate / 1024);
}
void AVTimeStamp::initVideoTimeStampParm(int fps, AVTimeStamp::PTSMode mode)
{
vMode = mode;
videoDuration = 1000000 / fps;
}
void AVTimeStamp::startTimeStamp()
```

```
audioTimeStamp = 0;
 videoTimeStamp = 0;
 startTime = av_gettime();
 int64_t AVTimeStamp::getAudioPts()
 if(aMode == PTS_RECTIFY)
 int64_t elapsed = av_gettime() - startTime;
 uint32_t offset = qAbs(elapsed - (audioTimeStamp + audioDuration));
 if(offset < (audioDuration * 0.5))
 audioTimeStamp \ += \ audioDuration;
 audioTimeStamp = elapsed;
 else
 audioTimeStamp = av_gettime() - startTime;
 return audioTimeStamp;
 int64_t AVTimeStamp::getVideoPts()
 if(vMode == PTS_RECTIFY)
 int64_t elapsed = av_gettime() - startTime;
 uint32_t offset = qAbs(elapsed - (videoTimeStamp + videoDuration));
 if(offset < (videoDuration * 0.5))</pre>
 videoTimeStamp += videoDuration;
 else
 videoTimeStamp = elapsed;
 else
 videoTimeStamp = av_gettime() - startTime;
 return videoTimeStamp;
4
                                                                          単报
评论1
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                                                                            数凸
    转发了
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```

分享一个用python带着原有格式拆分Excel表格到多张工作表的思路



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成都老小区决定自拆自建, 专家:第一个吃螃蟹的能提供 借鉴意义

齐鲁壹点 1473评论 4天前

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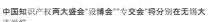
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潇湘晨报 5评论 2小时前

爸爸给孩子取名"宇凡", 护士手误多加一笔, 妈妈大喜:不改了









小雪妈妈育儿 347评论 6天

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晓瑜海棠 16评论 2小时前

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