

原 最简单的基于FFMPEG的转码程序

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本文介绍一个简单的基于FFmpeg的转码器。它可以将一种视频格式（包括封装格式和编码格式）转换为另一种视频格式。转码器在视音频编解码处理的程序中，属于一个比较复杂的东西。因为它结合了视频的解码和编码。一个视频播放器，一般只包含解码功能；一个视频编码工具，一般只包含编码功能；而一个视频转码器，则需要先对视频进行解码，然后再对视频进行编码，因而相当于解码器和编码器的结合。下图例举了一个视频的转码流程。输入视频的封装格式是FLV，视频编码标准是H.264，音频编码标准是AAC；输出视频的封装格式是AVI，视频编码标准是MPEG2，音频编码标准是MP3。从流程中可以看出，首先从输入视频中分离出视频码流和音频压缩码流，然后分别将视频码流和音频码流进行解码，获取到非压缩的像素数据/音频采样数据，接着将非压缩的像素数据/音频采样数据重新进行编码，获得重新编码后的视频码流和音频码流，最后将视频码流和音频码流重新封装成一个文件。

本文介绍的视频转码器正是使用FFMPEG类库从编程的角度实现了上述流程。该例子是从FFmpeg的例子改编的，平台是VC2010，类库版本是2014.5.6。

流程图（2014.9.29更新）

下面附两张使用FFmpeg转码视频的流程图。图中使用浅绿色标出了视频的编码、解码函数。从代码中可以看出，使用了AVFilter的不少东西，因此建议先学习AVFilter的内容后再看这个转码器的源代码。

PS：实际上，转码器不是一定依赖AVFilter的。因此打算有时间对这个转码器进行进一步的简化，使学习的人无需AVFilter的基础也可以理解转码器。

简单介绍一下流程中各个函数的意义：

open_input_file()：打开输入文件，并初始化相关的结构体。

open_output_file()：打开输出文件，并初始化相关的结构体。

init_filters()：初始化AVFilter相关的结构体。

av_read_frame()：从输入文件中读取一个AVPacket。

avcodec_decode_video2()：解码一个视频AVPacket（存储H.264等压缩码流数据）为AVFrame（存储YUV等非压缩的像素数据）。

avcodec_decode_audio4()：解码一个音频AVPacket（存储MP3等压缩码流数据）为AVFrame（存储PCM采样数据）。

filter_encode_write_frame()：编码一个AVFrame。

flush_encoder()：输入文件读取完毕后，输出编码器中剩余的AVPacket。

以上函数中open_input_file(), open_output_file(), init_filters()中的函数在其他文章中都有所叙述，在这里不再重复：

open_input_file()可参考：[100行代码实现最简单的基于FFMPEG+SDL的视频播放器（SDL1.x）](#)

open_output_file()可参考：[最简单的基于FFMPEG的视频编码器（YUV编码为H.264）](#)

init_filters()可参考：[最简单的基于FFmpeg的AVfilter例子（水印叠加）](#)

在这里介绍一下其中编码的函数filter_encode_write_frame()。filter_encode_write_frame()函数的流程如下图所示，它完成了视频/音频的编码功能。

PS：视频和音频的编码流程中除了编码函数avcodec_encode_video2()和avcodec_encode_audio2()不一样之外，其他部分几乎完全一样。

简单介绍一下filter_encode_write_frame()中各个函数的意义：

av_buffersrc_add_frame()：将解码后的AVFrame加入Filtergraph。

av_buffersink_get_buffer_ref()：从Filtergraph中取一个AVFrame。

avcodec_encode_video2()：编码一个视频AVFrame为AVPacket。

avcodec_encode_audio2()：编码一个音频AVFrame为AVPacket。

av_interleaved_write_frame()：将编码后的AVPacket写入文件。

代码

贴上代码

```

1.  /*
2.   *最简单的基于FFmpeg的转码器
3.   *Simplest FFmpeg Transcoder
4.   *
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10.  *
11.  *本程序实现了视频格式之间的转换。是一个最简单的视频转码程序。
12.  *
13.  */
14.
15. #include "stdafx.h"
16. extern "C"
17. {
18.     #include "libavcodec/avcodec.h"
19.     #include "libavformat/avformat.h"
20.     #include "libavfilter/avfiltergraph.h"
21.     #include "libavfilter/avcodec.h"
22.     #include "libavfilter/buffersink.h"
23.     #include "libavfilter/buffersrc.h"
24.     #include "libavutil/avutil.h"
25.     #include "libavutil/opt.h"
26.     #include "libavutil/pixdesc.h"
27. };
28.
29.
30.
31. static AVFormatContext *ifmt_ctx;
32. static AVFormatContext *ofmt_ctx;
33. typedef struct FilteringContext{
34.     AVFilterContext*buffersink_ctx;
35.     AVFilterContext*buffer_src_ctx;
36.     AVFilterGraph*filter_graph;
37. } FilteringContext;
38. static FilteringContext *filter_ctx;
39. static int open_input_file(const char *filename)
40. {
41.     int ret;
42.     unsigned int i;
43.     ifmt_ctx=NULL;
44.     if ((ret = avformat_open_input(&ifmt_ctx,filename, NULL, NULL)) < 0) {
45.         av_log(NULL, AV_LOG_ERROR, "Cannot openinput file\n");
46.         return ret;
47.     }
48.     if ((ret = avformat_find_stream_info(ifmt_ctx, NULL))< 0) {
49.         av_log(NULL, AV_LOG_ERROR, "Cannot findstream information\n");
50.         return ret;
51.     }
52.     for (i = 0; i < ifmt_ctx->nb_streams; i++) {
53.         AVStream*stream;
54.         AVCodecContext *codec_ctx;
55.         stream =ifmt_ctx->streams[i];
56.         codec_ctx =stream->codec;
57.         /* Reencode video & audio and remux subtitles etc. */
58.         if (codec_ctx->codec_type == AVMEDIA_TYPE_VIDEO
59.             ||codec_ctx->codec_type == AVMEDIA_TYPE_AUDIO) {
60.             /* Open decoder */
61.             ret =avcodec_open2(codec_ctx,
62.                               avcodec_find_decoder(codec_ctx->codec_id), NULL);
63.             if (ret < 0) {
64.                 av_log(NULL, AV_LOG_ERROR, "Failed toopen decoder for stream #%u\n", i);
65.                 return ret;
66.             }
67.         }
68.     }
69.     av_dump_format(ifmt_ctx, 0, filename, 0);
70.     return 0;
71. }
72. static int open_output_file(const char *filename)
73. {
74.     AVStream*out_stream;
75.     AVStream*in_stream;
76.     AVCodecContext*dec_ctx, *enc_ctx;
77.     AVCodec*encoder;
78.     int ret;
79.     unsigned int i;
80.     ofmt_ctx=NULL;
81.     avformat_alloc_output_context2(&ofmt_ctx, NULL, NULL, filename);
82.     if (!ofmt_ctx) {
83.         av_log(NULL, AV_LOG_ERROR, "Could notcreate output context\n");
84.         return AERROR_UNKNOWN;
85.     }
86.     for (i = 0; i < ifmt_ctx->nb_streams; i++) {
87.         out_stream= avformat_new_stream(ofmt_ctx, NULL);
88.         if (!out_stream) {
89.             av_log(NULL, AV_LOG_ERROR, "Failedallocating output stream\n");
90.             return AERROR_UNKNOWN;
91.         }

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92.     in_stream = ifmt_ctx->streams[i];
93.     dec_ctx = in_stream->codec;
94.     enc_ctx = out_stream->codec;
95.     if (dec_ctx->codec_type == AVMEDIA_TYPE_VIDEO
96.         || dec_ctx->codec_type == AVMEDIA_TYPE_AUDIO) {
97.         /* in this example, we choose transcoding to same codec */
98.         encoder = avcodec_find_encoder(dec_ctx->codec_id);
99.         /* In this example, we transcode to same properties (picture size,
100.          * sample rate etc.). These properties can be changed for output
101.          * streams easily using filters */
102.         if (dec_ctx->codec_type == AVMEDIA_TYPE_VIDEO) {
103.             enc_ctx->height = dec_ctx->height;
104.             enc_ctx->width = dec_ctx->width;
105.             enc_ctx->sample_aspect_ratio = dec_ctx->sample_aspect_ratio;
106.             /* take first format from list of supported formats */
107.             enc_ctx->pix_fmt = encoder->pix_fmts[0];
108.             /* video time_base can be set to whatever is handy and supported by encoder */
109.             enc_ctx->time_base = dec_ctx->time_base;
110.         } else {
111.             enc_ctx->sample_rate = dec_ctx->sample_rate;
112.             enc_ctx->channel_layout = dec_ctx->channel_layout;
113.             enc_ctx->channels = av_get_channel_layout_nb_channels(enc_ctx->channel_layout);
114.             /* take first format from list of supported formats */
115.             enc_ctx->sample_fmt = encoder->sample_fmts[0];
116.             AVRational time_base = {1, enc_ctx->sample_rate};
117.             enc_ctx->time_base = time_base;
118.         }
119.         /* Third parameter can be used to pass settings to encoder */
120.         ret = avcodec_open2(enc_ctx, encoder, NULL);
121.         if (ret < 0) {
122.             av_log(NULL, AV_LOG_ERROR, "Cannot open video encoder for stream %u\n", i);
123.             return ret;
124.         }
125.     } else if (dec_ctx->codec_type == AVMEDIA_TYPE_UNKNOWN) {
126.         av_log(NULL, AV_LOG_FATAL, "Elementary stream %d is of unknown type, cannot proceed\n", i);
127.         return AVERROR_INVALIDDATA;
128.     } else {
129.         /* if this stream must be remuxed */
130.         ret = avcodec_copy_context(ofmt_ctx->streams[i]->codec,
131.                                     ifmt_ctx->streams[i]->codec);
132.         if (ret < 0) {
133.             av_log(NULL, AV_LOG_ERROR, "Copying stream context failed\n");
134.             return ret;
135.         }
136.     }
137.     if (ofmt_ctx->oformat->flags & AVFMT_GLOBALHEADER)
138.         enc_ctx->flags |= CODEC_FLAG_GLOBAL_HEADER;
139. }
140. av_dump_format(ofmt_ctx, 0, filename, 1);
141. if (!(ofmt_ctx->oformat->flags & AVFMT_NOFILE)) {
142.     ret = avio_open(&ofmt_ctx->pb, filename, AVIO_FLAG_WRITE);
143.     if (ret < 0) {
144.         av_log(NULL, AV_LOG_ERROR, "Could not open output file '%s'", filename);
145.         return ret;
146.     }
147. }
148. /* init muxer, write output file header */
149. ret = avformat_write_header(ofmt_ctx, NULL);
150. if (ret < 0) {
151.     av_log(NULL, AV_LOG_ERROR, "Error occurred when opening output file\n");
152.     return ret;
153. }
154. return 0;
155. }
156. static int init_filter(FilteringContext* fctx, AVCodecContext *dec_ctx,
157.                       AVCodecContext *enc_ctx, const char *filter_spec)
158. {
159.     char args[512];
160.     int ret = 0;
161.     AVFilter* buffersrc = NULL;
162.     AVFilter* buffersink = NULL;
163.     AVFilterContext* buffersrc_ctx = NULL;
164.     AVFilterContext* buffersink_ctx = NULL;
165.     AVFilterInOut* outputs = avfilter_inout_alloc();
166.     AVFilterInOut* inputs = avfilter_inout_alloc();
167.     AVFilterGraph* filter_graph = avfilter_graph_alloc();
168.     if (!outputs || !inputs || !filter_graph) {
169.         ret = AVERROR(ENOMEM);
170.         goto end;
171.     }
172.     if (dec_ctx->codec_type == AVMEDIA_TYPE_VIDEO) {
173.         buffersrc = avfilter_get_by_name("buffer");
174.         buffersink = avfilter_get_by_name("buffersink");
175.         if (!buffersrc || !buffersink) {
176.             av_log(NULL, AV_LOG_ERROR, "filtering source or sink element not found\n");
177.             ret = AVERROR_UNKNOWN;
178.             goto end;
179.         }
180.         snprintf(args, sizeof(args),
181.                  "video_size=%dx%d:pix_fmt=%d:time_base=%d/%d:pixel_aspect=%d/%d",
182.                  dec_ctx->width, dec_ctx->height, dec_ctx->pix_fmt,

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183.         dec_ctx->time_base.num,dec_ctx->time_base.den,
184.         dec_ctx->sample_aspect_ratio.num,
185.         dec_ctx->sample_aspect_ratio.den);
186.     ret =avfilter_graph_create_filter(&buffersrc_ctx, buffersrc, "in",
187.     args, NULL, filter_graph);
188.     if (ret < 0) {
189.         av_log(NULL, AV_LOG_ERROR, "Cannotcreate buffer source\n");
190.         goto end;
191.     }
192.     ret =avfilter_graph_create_filter(&buffersink_ctx, buffersink, "out",
193.     NULL, NULL, filter_graph);
194.     if (ret < 0) {
195.         av_log(NULL, AV_LOG_ERROR, "Cannotcreate buffer sink\n");
196.         goto end;
197.     }
198.     ret =av_opt_set_bin(buffersink_ctx, "pix_fmts",
199.     (uint8_t*)&enc_ctx->pix_fmt, sizeof(enc_ctx->pix_fmt),
200.     AV_OPT_SEARCH_CHILDREN);
201.     if (ret < 0) {
202.         av_log(NULL, AV_LOG_ERROR, "Cannot setoutput pixel format\n");
203.         goto end;
204.     }
205. } else if(dec_ctx->codec_type == AVMEDIA_TYPE_AUDIO) {
206.     buffersrc = avfilter_get_by_name("abuffer");
207.     buffersink= avfilter_get_by_name("abuffersink");
208.     if (!buffersrc || !buffersink) {
209.         av_log(NULL, AV_LOG_ERROR, "filteringsource or sink element not found\n");
210.         ret =AERROR_UNKNOWN;
211.         goto end;
212.     }
213.     if (!dec_ctx->channel_layout)
214.         dec_ctx->channel_layout =
215.             av_get_default_channel_layout(dec_ctx->channels);
216.     _snprintf(args, sizeof(args),
217.         "time_base=%d/%d:sample_rate=%d:sample_fmt=%s:channel_layout=0x%I64x",
218.         dec_ctx->time_base.num, dec_ctx->time_base.den,dec_ctx->sample_rate,
219.         av_get_sample_fmt_name(dec_ctx->sample_fmt),
220.         dec_ctx->channel_layout);
221.     ret =avfilter_graph_create_filter(&buffersrc_ctx, buffersrc, "in",
222.     args, NULL, filter_graph);
223.     if (ret < 0) {
224.         av_log(NULL, AV_LOG_ERROR, "Cannotcreate audio buffer source\n");
225.         goto end;
226.     }
227.     ret =avfilter_graph_create_filter(&buffersink_ctx, buffersink, "out",
228.     NULL, NULL, filter_graph);
229.     if (ret < 0) {
230.         av_log(NULL, AV_LOG_ERROR, "Cannotcreate audio buffer sink\n");
231.         goto end;
232.     }
233.     ret = av_opt_set_bin(buffersink_ctx, "sample_fmts",
234.     (uint8_t*)&enc_ctx->sample_fmt, sizeof(enc_ctx->sample_fmt),
235.     AV_OPT_SEARCH_CHILDREN);
236.     if (ret < 0) {
237.         av_log(NULL, AV_LOG_ERROR, "Cannot setoutput sample format\n");
238.         goto end;
239.     }
240.     ret =av_opt_set_bin(buffersink_ctx, "channel_layouts",
241.     (uint8_t*)&enc_ctx->channel_layout,
242.     sizeof(enc_ctx->channel_layout),AV_OPT_SEARCH_CHILDREN);
243.     if (ret < 0) {
244.         av_log(NULL, AV_LOG_ERROR, "Cannot setoutput channel layout\n");
245.         goto end;
246.     }
247.     ret =av_opt_set_bin(buffersink_ctx, "sample_rates",
248.     (uint8_t*)&enc_ctx->sample_rate, sizeof(enc_ctx->sample_rate),
249.     AV_OPT_SEARCH_CHILDREN);
250.     if (ret < 0) {
251.         av_log(NULL, AV_LOG_ERROR, "Cannot setoutput sample rate\n");
252.         goto end;
253.     }
254. } else {
255.     ret =AERROR_UNKNOWN;
256.     goto end;
257. }
258. /* Endpoints for the filter graph. */
259. outputs->name      =av_strdup("in");
260. outputs->filter_ctx = buffersrc_ctx;
261. outputs->pad_idx    = 0;
262. outputs->next        = NULL;
263. inputs->name        = av_strdup("out");
264. inputs->filter_ctx  = buffersink_ctx;
265. inputs->pad_idx      = 0;
266. inputs->next         = NULL;
267. if (!outputs->name || !inputs->name) {
268.     ret =AERROR(ENOMEM);
269.     goto end;
270. }
271. if ((ret = avfilter_graph_parse_ptr(filter_graph,filter_spec,
272.     &inputs, &outputs, NULL)) < 0)
273.     goto end;

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274.     if ((ret = avfilter_graph_config(filter_graph, NULL)) < 0)
275.         goto end;
276.     /* Fill FilteringContext */
277.     fctx->buffersrc_ctx = buffersrc_ctx;
278.     fctx->buffersink_ctx = buffersink_ctx;
279.     fctx->filter_graph = filter_graph;
280. end:
281.     avfilter_inout_free(&inputs);
282.     avfilter_inout_free(&outputs);
283.     return ret;
284. }
285. static int init_filters(void)
286. {
287.     const char*filter_spec;
288.     unsigned int i;
289.     int ret;
290.     filter_ctx = (FilteringContext *)av_malloc_array(ifmt_ctx->nb_streams, sizeof(*filter_ctx));
291.     if (!filter_ctx)
292.         return AVERROR(ENOMEM);
293.     for (i = 0; i < ifmt_ctx->nb_streams; i++) {
294.         filter_ctx[i].buffersrc_ctx = NULL;
295.         filter_ctx[i].buffersink_ctx = NULL;
296.         filter_ctx[i].filter_graph = NULL;
297.         if (!(ifmt_ctx->streams[i]->codec->codec_type == AVMEDIA_TYPE_AUDIO
298.              || ifmt_ctx->streams[i]->codec->codec_type == AVMEDIA_TYPE_VIDEO))
299.             continue;
300.         if (ifmt_ctx->streams[i]->codec->codec_type == AVMEDIA_TYPE_VIDEO)
301.             filter_spec = "null"; /* passthrough (dummy) filter for video */
302.         else
303.             filter_spec = "anull"; /* passthrough (dummy) filter for audio */
304.         ret = init_filter(&filter_ctx[i], ifmt_ctx->streams[i]->codec,
305.                          ofmt_ctx->streams[i]->codec, filter_spec);
306.         if (ret)
307.             return ret;
308.     }
309.     return 0;
310. }
311. static int encode_write_frame(AVFrame *filt_frame, unsigned int stream_index, int*got_frame) {
312.     int ret;
313.     int got_frame_local;
314.     AVPacketenc_pkt;
315.     int (*enc_func)(AVCodecContext *, AVPacket *, const AVFrame *, int*) =
316.         (ifmt_ctx->streams[stream_index]->codec->codec_type ==
317.          AVMEDIA_TYPE_VIDEO) ? avcodec_encode_video2 : avcodec_encode_audio2;
318.     if (!got_frame)
319.         got_frame = &got_frame_local;
320.     av_log(NULL, AV_LOG_INFO, "Encoding frame\n");
321.     /* encode filtered frame */
322.     enc_pkt.data = NULL;
323.     enc_pkt.size = 0;
324.     av_init_packet(&enc_pkt);
325.     ret = enc_func(ofmt_ctx->streams[stream_index]->codec, &enc_pkt,
326.                   filt_frame, got_frame);
327.     av_frame_free(&filt_frame);
328.     if (ret < 0)
329.         return ret;
330.     if (!(*got_frame))
331.         return 0;
332.     /* prepare packet for muxing */
333.     enc_pkt.stream_index = stream_index;
334.     enc_pkt.dts = av_rescale_q_rnd(enc_pkt.dts,
335.                                    ofmt_ctx->streams[stream_index]->codec->time_base,
336.                                    ofmt_ctx->streams[stream_index]->time_base,
337.                                    (AVRounding)(AV_ROUND_NEAR_INF|AV_ROUND_PASS_MINMAX));
338.     enc_pkt.pts = av_rescale_q_rnd(enc_pkt.pts,
339.                                    ofmt_ctx->streams[stream_index]->codec->time_base,
340.                                    ofmt_ctx->streams[stream_index]->time_base,
341.                                    (AVRounding)(AV_ROUND_NEAR_INF|AV_ROUND_PASS_MINMAX));
342.     enc_pkt.duration = av_rescale_q(enc_pkt.duration,
343.                                     ofmt_ctx->streams[stream_index]->codec->time_base,
344.                                     ofmt_ctx->streams[stream_index]->time_base);
345.     av_log(NULL, AV_LOG_DEBUG, "Muxing frame\n");
346.     /* mux encoded frame */
347.     ret = av_interleaved_write_frame(ofmt_ctx, &enc_pkt);
348.     return ret;
349. }
350. static int filter_encode_write_frame(AVFrame *frame, unsigned int stream_index)
351. {
352.     int ret;
353.     AVFrame*filt_frame;
354.     av_log(NULL, AV_LOG_INFO, "Pushing decoded frame to filters\n");
355.     /* push the decoded frame into the filtergraph */
356.     ret = av_buffersrc_add_frame_flags(filter_ctx[stream_index].buffersrc_ctx,
357.                                       frame, 0);
358.     if (ret < 0) {
359.         av_log(NULL, AV_LOG_ERROR, "Error while feeding the filtergraph\n");
360.         return ret;
361.     }
362.     /* pull filtered frames from the filtergraph */
363.     while (1) {
364.         filt_frame = av_frame_alloc();

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365.     if (!filt_frame) {
366.         ret = AVERROR(ENOMEM);
367.         break;
368.     }
369.     av_log(NULL, AV_LOG_INFO, "Pulling filtered frame from filters\n");
370.     ret = av_buffersink_get_frame(filter_ctx[stream_index].buffersink_ctx,
371.         filt_frame);
372.     if (ret < 0) {
373.         /* if nomore frames for output - returns AVERROR(EAGAIN)
374.          * if flushed and no more frames for output - returns AVERROR_EOF
375.          * rewrite retcode to 0 to show it as normal procedure completion
376.          */
377.         if (ret == AVERROR(EAGAIN) || ret == AVERROR_EOF)
378.             ret = 0;
379.         av_frame_free(&filt_frame);
380.         break;
381.     }
382.     filt_frame->pict_type = AV_PICTURE_TYPE_NONE;
383.     ret = encode_write_frame(filt_frame, stream_index, NULL);
384.     if (ret < 0)
385.         break;
386. }
387. return ret;
388. }
389. static int flush_encoder(unsigned int stream_index)
390. {
391.     int ret;
392.     int got_frame;
393.     if (!ofmt_ctx->streams[stream_index]->codec->capabilities &
394.         CODEC_CAP_DELAY)
395.         return 0;
396.     while (1) {
397.         av_log(NULL, AV_LOG_INFO, "Flushing stream #%u encoder\n", stream_index);
398.         ret = encode_write_frame(NULL, stream_index, &got_frame);
399.         if (ret < 0)
400.             break;
401.         if (!got_frame)
402.             return 0;
403.     }
404.     return ret;
405. }
406.
407. int_tmain(int argc, _TCHAR* argv[])
408. {
409.     int ret;
410.     AVPacket packet;
411.     AVFrame *frame = NULL;
412.     enum AVMediaType type;
413.     unsigned int stream_index;
414.     unsigned int i;
415.     int got_frame;
416.     int (*dec_func)(AVCodecContext *, AVFrame *, int *, const AVPacket*);
417.     if (argc != 3) {
418.         av_log(NULL, AV_LOG_ERROR, "Usage: %s<input file> <output file>\n", argv[0]);
419.         return 1;
420.     }
421.     av_register_all();
422.     avfilter_register_all();
423.     if ((ret = open_input_file(argv[1])) < 0)
424.         goto end;
425.     if ((ret = open_output_file(argv[2])) < 0)
426.         goto end;
427.     if ((ret = init_filters()) < 0)
428.         goto end;
429.     /* read all packets */
430.     while (1) {
431.         if ((ret = av_read_frame(ifmt_ctx, &packet)) < 0)
432.             break;
433.         stream_index = packet.stream_index;
434.         type = ifmt_ctx->streams[stream_index]->codec->codec_type;
435.         av_log(NULL, AV_LOG_DEBUG, "Demuxer gave frame of stream_index %u\n",
436.             stream_index);
437.         if (filter_ctx[stream_index].filter_graph) {
438.             av_log(NULL, AV_LOG_DEBUG, "Going to encode & filter the frame\n");
439.             frame = av_frame_alloc();
440.             if (!frame) {
441.                 ret = AVERROR(ENOMEM);
442.                 break;
443.             }
444.             packet.dts = av_rescale_q_rnd(packet.dts,
445.                 ifmt_ctx->streams[stream_index]->time_base,
446.                 ifmt_ctx->streams[stream_index]->codec->time_base,
447.                 (AVRounding)(AV_ROUND_NEAR_INF|AV_ROUND_PASS_MINMAX));
448.             packet.pts = av_rescale_q_rnd(packet.pts,
449.                 ifmt_ctx->streams[stream_index]->time_base,
450.                 ifmt_ctx->streams[stream_index]->codec->time_base,
451.                 (AVRounding)(AV_ROUND_NEAR_INF|AV_ROUND_PASS_MINMAX));
452.             dec_func = (type == AVMEDIA_TYPE_VIDEO) ? avcodec_decode_video2 :
453.                 avcodec_decode_audio4;
454.             ret = dec_func(ifmt_ctx->streams[stream_index]->codec, frame,
455.                 &got_frame, &packet);

```

```

456.         if (ret < 0) {
457.             av_frame_free(&frame);
458.             av_log(NULL, AV_LOG_ERROR, "Decodingfailed\n");
459.             break;
460.         }
461.         if (got_frame) {
462.             frame->pts = av_frame_get_best_effort_timestamp(frame);
463.             ret= filter_encode_write_frame(frame, stream_index);
464.             av_frame_free(&frame);
465.             if (ret< 0)
466.                 goto end;
467.         } else {
468.             av_frame_free(&frame);
469.         }
470.     } else {
471.         /* remux this frame without reencoding */
472.         packet.dts = av_rescale_q_rnd(packet.dts,
473.             ifmt_ctx->streams[stream_index]->time_base,
474.             ofmt_ctx->streams[stream_index]->time_base,
475.             (AVRounding)(AV_ROUND_NEAR_INF|AV_ROUND_PASS_MINMAX));
476.         packet.pts = av_rescale_q_rnd(packet.pts,
477.             ifmt_ctx->streams[stream_index]->time_base,
478.             ofmt_ctx->streams[stream_index]->time_base,
479.             (AVRounding)(AV_ROUND_NEAR_INF|AV_ROUND_PASS_MINMAX));
480.         ret =av_interleaved_write_frame(ofmt_ctx, &packet);
481.         if (ret < 0)
482.             goto end;
483.     }
484.     av_free_packet(&packet);
485. }
486. /* flush filters and encoders */
487. for (i = 0; i < ifmt_ctx->nb_streams; i++) {
488.     /* flush filter */
489.     if (!filter_ctx[i].filter_graph)
490.         continue;
491.     ret =filter_encode_write_frame(NULL, i);
492.     if (ret < 0) {
493.         av_log(NULL, AV_LOG_ERROR, "Flushingfilter failed\n");
494.         goto end;
495.     }
496.     /* flush encoder */
497.     ret = flush_encoder(i);
498.     if (ret < 0) {
499.         av_log(NULL, AV_LOG_ERROR, "Flushingencoder failed\n");
500.         goto end;
501.     }
502. }
503. av_write_trailer(ofmt_ctx);
504. end:
505. av_free_packet(&packet);
506. av_frame_free(&frame);
507. for (i = 0; i < ifmt_ctx->nb_streams; i++) {
508.     avcodec_close(ifmt_ctx->streams[i]->codec);
509.     if (ofmt_ctx && ofmt_ctx->nb_streams > i && ofmt_ctx->streams[i] &&ofmt_ctx->streams[i]->codec)
510.         avcodec_close(ofmt_ctx->streams[i]->codec);
511.     if(filter_ctx && filter_ctx[i].filter_graph)
512.         avfilter_graph_free(&filter_ctx[i].filter_graph);
513. }
514. av_free(filter_ctx);
515. avformat_close_input(&ifmt_ctx);
516. if (ofmt_ctx &&!(ofmt_ctx->oformat->flags & AVFMT_NOFILE))
517.     avio_close(ofmt_ctx->pb);
518. avformat_free_context(ofmt_ctx);
519. if (ret < 0)
520.     av_log(NULL, AV_LOG_ERROR, "Erroroccurred\n");
521. return (ret? 1:0);
522. }

```

程序运行截图：

□

默认情况下运行程序，会将“cuc_ieschool.ts”转换为“cuc_ieschool.avi”。调试的时候，可以修改“配置属性->调试->命令参数”中的参数，即可改变转码的输入输出文件。

□

工程下载地址（VC2010）：<http://download.csdn.net/detail/leixiaohua1020/7394649>

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