## ສ ffmpeg源码分析:transcode()函数

2013年09月20日 17:34:05 阅读数:6227

## 还是先看一下主函数吧:(省略了很多无关大雅的代码)

```
[cpp] 📳 🔝
     int main(int argc, char **argv)
2.
     {
3.
         OptionsContext o = { 0 };
     int64_t ti;
4.
5.
    //与命令行分析有关的结构的初始化,下面不再罗嗦
6.
         reset options(&o, 0);
8.
q
         //设置日志级别
    av_log_set_flags(AV_LOG_SKIP_REPEATED);
10.
11.
         parse_loglevel(argc, argv, options);
12.
13.
         if (argc > 1 && !strcmp(argv[1], "-d")) {
14.
     run_as_daemon = 1;
15.
             av_log_set_callback(log_callback_null);
16.
            argc--;
17.
            argv++;
18.
19.
     //注册组件们
20.
21.
         avcodec_register_all();
     #if CONFIG AVDEVICE
22.
23.
         avdevice_register_all();
24.
     #endif
25.
     #if CONFIG_AVFILTER
26.
        avfilter_register_all();
27.
28.
     av_register_all();
         //初始化网络,windows下需要
29.
30.
     avformat_network_init();
31.
32.
     show banner();
33.
     term_init();
34.
35.
     //分析命令行输入的参数们
36.
37.
         parse_options(&o, argc, argv, options, opt_output_file);
38.
39.
         //文件的转换就在此函数中发生
    if (transcode(output_files, nb_output_files, input_files, nb_input_files)< 0)</pre>
40.
41.
            exit_program(1);
42.
43.
         exit_program(0);
44.
        return 0;
45. }
```

下面是transcode()函数,转换就发生在它里面.不废话,看注释吧,应很详细了

```
[cpp] 📳 📑
     static int transcode(
1.
      OutputFile *output_files,//输出文件数组
3.
            int nb_output_files,//输出文件的数量
4.
           InputFile *input_files,//输入文件数组
5.
            int nb_input_files)//输入文件的数量
6.
     {
        int ret, i;
     AVFormatContext *is, *os;
8.
9.
        OutputStream *ost;
10.
     InputStream *ist;
11.
        uint8 t *no packet;
     int no_packet_count = 0;
12.
        int64_t timer_start;
13.
     int key;
14.
15.
    if (!(no_packet = av_mallocz(nb_input_files)))
16.
17.
            exit_program(1);
18.
         //设置编码参数,打开所有输出流的编码器,打开所有输入流的解码器,写入所有输出文件的文件头,于是准备好了
19.
20.
     ret = transcode_init(output_files, nb_output_files, input_files,nb_input_files);
21.
         if (ret < 0)
22.
        goto fail;
23.
24.
         if (!using_stdin){
            av_log(NULL, AV_LOG_INFO, "Press [q] to stop, [?] for help\n");
25.
26.
```

```
28.
         timer_start = av_gettime();
 29.
 30.
       //循环,直到收到系统信号才退出
 31.
          for (; received_sigterm == 0;)
 32.
 33.
              int file_index, ist_index;
 34.
              AVPacket pkt;
 35.
              int64 t ipts min;
              double opts min;
 36.
 37.
              int64_t cur_time = av_gettime();
 38.
 39.
              ipts min = INT64 MAX;
 40.
              opts min = 1e100;
 41.
              /* if 'q' pressed, exits */
 42
              if (!using_stdin)
 43.
 44.
                  //先查看用户按下了什么键,跟据键做出相应的反应
 45.
                  static int64_t last_time;
 46.
                  if (received_nb_signals)
 47.
                     break;
                   /* read_key() returns 0 on EOF */
 48.
 49.
                  if (cur_time - last_time >= 100000 && !run_as_daemon){
 50.
                     key = read key();
 51.
                     last time = cur time;
 52.
                  }else{
 53.
                       </span>.....
       <span>
 54.
 55.
 56.
              /st select the stream that we must read now by looking at the
 57.
               smallest output pts */
 58.
              //下面这个循环的目的是找一个最小的输出pts(也就是离当前最近的)的输出流
 59.
              file_index = -1;
 60.
              for (i = 0; i < nb_output_streams; i++){</pre>
 61.
                  OutputFile *of;
                  int64_t ipts;
 62.
 63.
                  double opts;
 64.
                  ost = &output streams[i];//循环每一个输出流
 65.
                  of = &output_files[ost->file_index];//输出流对应的输出文件
                  os = output_files[ost->file_index].ctx;//输出流对应的FormatContext
 66.
 67.
                  ist = &input streams[ost->source index];//输出流对应的输入流
 68.
                  if (ost->is_past_recording_time || //是否过了录制时间?(可能用户指定了一个录制时间段)
 69.
 70.
                         no_packet[ist->file_index] | //对应的输入流这个时间内没有数据?
 71.
                          (os->pb && avio_tell(os->pb) >= of->limit_filesize))//是否超出了录制范围(也是用户指定的)
 72.
                      continue;//是的,符合上面某一条,那么再看下一个输出流吧
 73.
 74.
                  //判断当前输入流所在的文件是否可以使用(我也不很明白)
 75.
                  opts = ost->st->pts.val * av_q2d(ost->st->time_base);
                  ipts = ist->pts;
 76.
 77.
                  if (!input_files[ist->file_index].eof_reached)
 78.
                     if (ipts < ipts_min){</pre>
                          //每找到一个pts更小的输入流就记录下来,这样循环完所有的输出流时就找到了
 79.
                         //pts最小的输入流,及输入文件的序号
 80.
 81.
                          ipts min = ipts;
                         if (input sync)
 82.
                             file index = ist->file index;
 83.
 84.
                      if (opts < opts_min){</pre>
 85.
 86.
                         opts min = opts;
 87.
                         if (!input_sync)
 88.
                             file_index = ist->file_index;
 89.
 90.
 91.
                  //难道下面这句话的意思是:如果当前的输出流已接收的帧数,超出用户指定的输出最大帧数时,
 92.
                  //则当前输出流所属的输出文件对应的所有输出流,都算超过了录像时间?
 93.
 94.
                  if (ost->frame_number >= ost->max_frames){
 95.
                     int j;
                      for (i = 0: i < of->ctx->nb streams: i++)
 96.
                         output_streams[of->ost_index + j].is_past_recording_time = 1;
 97.
 98.
                      continue:
 99.
100.
101.
              /* if none, if is finished */
102.
              if (file_index < 0) {</pre>
103
                  //如果没有找到合适的输入文件
104.
                  if (no_packet_count){
105.
                      //如果是因为有的输入文件暂时得不到数据,则还不算是结束
106.
                     no_packet_count = 0;
107.
                     memset(no packet, 0, nb input files);
108.
                     usleep(10000);
109.
                     continue;
110.
                  //全部转换完成了,跳出大循环
111.
112.
                  break:
113.
              }
114.
              //从找到的输入文件中读出一帧(可能是音频也可能是视频),并放到fifo队列中
115.
116
              is = input_files[file_index].ctx;
117.
              ret = av_read_frame(is, &pkt);
              if (rot -- AVERROR(EAGAIN)) }
```

```
- AVENNUN(LAUMIN//) \
119.
                   //此时发生了暂时没数据的情况
120.
                   no packet[file index] = 1;
121.
                   no packet count++:
122.
                   continue;
123.
               }
124.
125.
               //下文判断是否有输入文件到最后了
126.
               if (ret < 0){
127
                   input_files[file_index].eof_reached = 1;
128.
                   if (opt_shortest)
129.
                       break;
130.
131.
                       continue:
132.
133.
134.
               no packet count = 0;
               memset(no_packet, 0, nb_input_files);
135.
136.
137.
               if (do pkt dump){
                  av_pkt_dump_log2(NULL, AV_LOG_DEBUG, &pkt, do_hex_dump,
138.
139.
                           is->streams[pkt.stream_index]);
140.
141.
               /* the following test is needed in case new streams appear
142.
               dynamically in stream : we ignore them */
143.
               //如果在输入文件中遇到一个忽然冒出的流,那么我们不鸟它
144
               if (pkt.stream_index >= input_files[file_index].nb_streams)
                   goto discard_packet;
145.
146.
               //取得当前获得的帧对应的输入流
147.
148.
               ist_index = input_files[file_index].ist_index + pkt.stream_index;
149.
               ist = &input streams[ist index];
               if (ist->discard)
150.
151.
                   goto discard packet;
152.
               //重新鼓捣一下帧的时间戳
153.
               if (pkt.dts != AV NOPTS VALUE)
154.
155.
                   pkt.dts += av_rescale_q(input_files[ist->file_index].ts_offset,
156.
                          AV_TIME_BASE_Q, ist->st->time_base);
157.
               if (pkt.pts != AV_NOPTS_VALUE)
158.
                   pkt.pts += av_rescale_q(input_files[ist->file_index].ts_offset,
                           AV_TIME_BASE_Q, ist->st->time_base);
159.
160.
161.
               if (pkt.pts != AV_NOPTS_VALUE)
                  pkt.pts *= ist->ts_scale;
162.
163.
               if (pkt.dts != AV NOPTS VALUE)
                pkt.dts *= ist->ts scale;
164.
165.
166.
               if (pkt.dts != AV NOPTS VALUE && ist->next pts != AV NOPTS VALUE
                       && (is->iformat->flags & AVFMT TS DISCONT))
167.
168
169.
                   int64_t pkt_dts = av_rescale_q(pkt.dts, ist->st->time_base,
170
                          AV_TIME_BASE_Q);
171.
                   int64_t delta = pkt_dts - ist->next_pts;
                   if ((delta < -1LL * dts_delta_threshold * AV_TIME_BASE</pre>
172
173.
                           || (delta > 1LL * dts_delta_threshold * AV_TIME_BASE
                                   && ist->st->codec->codec_type
174.
175.
                                           != AVMEDIA_TYPE_SUBTITLE)
176.
                          || pkt_dts + 1 < ist->pts) && !copy_ts)
177.
178.
                      input files[ist->file index].ts offset -= delta;
179.
                       av_log( NULL, AV_LOG_DEBUG,
                               "timestamp discontinuity %"PRId64", new offset= %"PRId64"\n",
180.
                               delta, input_files[ist->file_index].ts_offset);
181.
                       pkt.dts -= av_rescale_q(delta, AV_TIME_BASE_Q, ist->st->time_base);
182.
                       if (pkt.pts != AV NOPTS VALUE)
183.
184.
                           pkt.pts -= av_rescale_q(delta, AV_TIME_BASE_Q, ist->st->time_base);
185
186.
187
188.
               //把这一帧转换并写入到输出文件中
189.
               if (output_packet(ist, output_streams, nb_output_streams, &pkt) < 0){</pre>
190.
                   av_log(NULL, AV_LOG_ERROR,
191.
                           "Error while decoding stream #%d:%d\n",
192.
                           ist->file index, ist->st->index);
193.
                   if (exit_on_error)
194.
                     exit program(1);
195.
                   av free packet(&pkt):
196.
                   continue:
197.
               1
198.
199.
       discard packet:
200.
               av_free_packet(&pkt);
201.
202.
               /* dump report by using the output first video and audio streams */
203.
               print_report(output_files, output_streams, nb_output_streams, 0,
204.
                      timer_start, cur_time);
205.
206.
           //文件处理完了,把缓冲中剩余的数据写到输出文件中
207.
           for (i = 0; i < nb input streams; i++){}
208.
209
               ist = &input streams[il:
```

```
210.
              if (ist->decoding_needed){
211.
                  output_packet(ist, output_streams, nb_output_streams, NULL);
212.
213.
214.
       flush_encoders(output_streams, nb_output_streams);
215.
216.
       term_exit();
217.
218.
      //为输出文件写文件尾(有的不需要).
219.
           for (i = 0; i < nb output files; i++){
220.
          os = output files[i].ctx:
221.
               av_write_trailer(os);
222.
223.
224.
       /* dump report by using the first video and audio streams */
225.
          print_report(output_files, output_streams, nb_output_streams, 1,
226.
                  timer_start, av_gettime());
227.
228.
       //关闭所有的编码器
229.
           for (i = 0; i < nb_output_streams; i++){</pre>
230.
              ost = &output_streams[i];
231.
               if (ost->encoding needed){
232.
                  av_freep(&ost->st->codec->stats_in);
233.
                  avcodec_close(ost->st->codec);
234.
              }
235.
       #if CONFIG AVFILTER
             avfilter_graph_free(&ost->graph);
236.
237.
       #endif
238.
       }
239.
       //关闭所有的解码器
240.
241.
           for (i = 0; i < nb_input_streams; i++){</pre>
242.
            ist = &input_streams[i];
243.
               if (ist->decoding_needed){
244.
                  avcodec_close(ist->st->codec);
245.
              }
246.
247.
248.
       /* finished ! */
249.
           ret = 0:
250.
251.
           fail: av freep(&bit buffer):
       av_freep(&no_packet);
252.
253.
254.
       if (output_streams) {
255.
               for (i = 0; i < nb_output_streams; i++) {</pre>
256.
                  ost = &output_streams[i];
                   if (ost) {
257.
258.
                      if (ost->stream_copy)
259.
                          av_freep(&ost->st->codec->extradata);
260.
                       if (ost->logfile){
261.
                          fclose(ost->logfile);
                          ost->logfile = NULL;
262.
263.
                      av fifo free(ost->fifo); /* works even if fifo is not
264.
265.
                       initialized but set to zero */
266.
                       av freep(&ost->st->codec->subtitle header);
267
                       av_free(ost->resample_frame.data[0]);
268.
                       av_free(ost->forced_kf_pts);
269.
                       if (ost->video_resample)
270.
                          sws_freeContext(ost->img_resample_ctx);
271.
                       swr_free(&ost->swr);
272.
                       av_dict_free(&ost->opts);
273.
274.
275.
276.
          return ret;
277.
```

原文地址: http://blog.csdn.net/niu\_gao/article/details/7175421

文章标签: ffmpeg 源代码 transcode 个人分类: FFMPEG

所属专栏: 开源多媒体项目源代码分析 FFmpeg

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