

Linux3.0.8平台搭建移植文档——alsa声卡驱动移植

1)添加 ac97控制器时钟

ac97.c 文件对应着 s5pv210的 ac97控制器驱动程序,并且采用了 platform 机制,在 platform_driver 的 probe 回调函数中会执行 clk_get(&pdev->dev, "ac97");函数获取 ac97的时钟,因此我们需要进行 ac97时钟的添加。

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钟,因此我们需要进行 ac97时钟的添加。
#vi arch/arm/mach-s5pv210/clock.c
在 init clocks off 数组开头添加如下成员:
                       = "ac97",
          . name
          . id
                       = -1,
                       = &clk pclk psys. clk,
          .parent
                       = s5pv210 clk ip3 ctr1,
          .enable
                       = (1 << 1),
          .ctrlbit
     },
2) 修改 wm9713 codec 驱动
   #vi sound/soc/codecs/wm9713.c
   Stepl: 修改 wm9713 set pl1()函数中 ac98 write 函数参数:
static int wm9713_set_pll(struct snd_soc_codec *codec,int pll_id, unsigned int freq_in, unsigned int freq_out)
{
   /* turn PLL on and select as source */
   reg = ac97_read(codec, AC97_EXTENDED_MID);
   ac97_write(codec, AC97_EXTENDED_MID, reg & 0x7dff); //修改参数
   reg = ac97 read(codec, AC97 HANDSET RATE);
   ac97 write(codec, AC97 HANDSET RATE, reg & 0xff7f);
   wm9713-p11 in = freq in;
   /* wait 10ms AC97 link frames for the link to stabilise */
   schedule_timeout_interruptible(msecs_to_jiffies(10));
   return 0;
     Step2: 在 wm9713_pcm_hw_params()函数前面,添加以下函数:
     static int wm9713 hifi hw params(struct snd pcm substream *substream,
                          struct snd_pcm_hw_params *params,
                          struct snd soc dai *dai)
     {
          struct snd_soc_codec *codec = dai->codec;
          ac97 write(codec, AC97 POWERDOWN, 0x0000);
          ac97_write(codec, AC97_PHONE, 0x0808);
          /* hcj for dbg */
          ac97_write(codec, AC97_EXTENDED_MID, 0x7803);
          ac97 write(codec, AC97 EXTENDED MSTATUS, 0xb990);
```

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```
ac97_write(codec, AC97_MASTER, 0x8080);
      ac97 write (codec, AC97 HEADPHONE, 0x0606);
      ac97_write(codec, AC97_REC_GAIN, 0x00aa);
 #ifdef CONFIG SOUND WM9713 INPUT STREAM MIC
      ac97_write(codec, 0x5c, 0x0002);
      ac97_write(codec, AC97_LINE, 0x0068);
      ac97_write(codec, AC97_VIDEO, 0xfe00);
 #else
      ac97 write(codec, AC97 VIDEO, 0xd612);
 #endif
      return 0;
Step3: 在 wm9713 dai ops hifi 结构体成员初始化中添加如下内容:
 static struct snd soc dai ops wm9713 dai ops hifi = {
                      = wm9713 hifi hw params, //add by sunpluss
      .hw params
                    = ac97_hifi_prepare,
      .prepare
                       = wm9713_set_dai_clkdiv,
      .set_clkdiv
                    = wm9713 set dai p11,
      .set pll
};
Step4: 修改 wm9713_reset()函数如下:
int wm9713_reset(struct snd_soc_codec *codec, int try_warm)
 {
 /*
        if (try_warm && soc_ac97_ops.warm_reset) {
           soc_ac97_ops.warm_reset(codec->ac97);
           if (ac97\_read(codec, 0) == wm9713\_reg[0])
                 return 1;
      soc_ac97_ops.reset(codec->ac97);
      if (soc_ac97_ops.warm_reset)
           soc_ac97_ops.warm_reset(codec->ac97);
      if (ac97_read(codec, 0) != wm9713_reg[0])
           return -EIO;
 */
      if (try_warm && soc_ac97_ops.warm_reset) {
            while (ac97_read(codec, 0) != wm9713_reg[0])
                 soc_ac97_ops.warm_reset(codec->ac97);
           //if (ac97\_read(codec, 0) == wm9713\_reg[0])
                 return 1;
```



```
return 0;
     }
Step5: 修改函数 wm9713 set bias level()如下:
static int wm9713_set_bias_level(struct snd_soc_codec *codec, enum snd_soc_bias_level level)
{
   case SND_SOC_BIAS_OFF:
   /* disable everything including AC link */
    ac97 write(codec, AC97 EXTENDED MID, 0x7fff);
    ac97 write(codec, AC97 EXTENDED MSTATUS, Oxffff);
    ac97 write(codec, AC97_POWERDOWN, Oxfffff);
}
3) make menuconfig 配置 alsa 声卡驱动
     #vi make menuconfig
    Device Drivers --->
         [*]Sound card support --->
                  Advanced Linux Sound Architecture --->//选中该选项支持 alsa 总线驱动
             < >
                   Open Sound System (DEPRECATED) --->
进入 Advanced Linux Sound Architecture 选项配置如下:
--- Advanced Linux Sound Architecture
< >
      Sequencer support
<*>
     OSS Mixer API
<*>
     OSS PCM (digital audio) API
[*]
       OSS PCM (digital audio) API - Include plugin system
< >
     HR-timer backend support
Dynamic device file minor numbers
[*]
      Support old ALSA API
[*]
     Verbose procfs contents
Verbose printk
[ ]
     Debug
[ ]
     Generic sound devices --->
ARM sound devices --->
(*)
     ALSA for SoC audio support --->
           --- ALSA for SoC audio support
          Support LZO compression for register caches
          <*>
                ASoC support for Samsung
          < >
                SoC I2S Audio support for WM8580 on SMDK
          <*>
                SoC AC97 Audio support for SMDK with WM9713
          < >
                SoC S/PDIF Audio support for SMDK
          < >
                SoC PCM Audio support for WM8580 on SMDK
                 Build all ASoC CODEC drivers
```



4) make

将在arch/arm/boot/下生成编译好的可执行程序**zImage**下载到开发板即可,成功移植后会在看到1s /dev/dsp这个设备文件,执行命令"echo /dev/zero > /dev/dsp"可以听到"吱"的声音,也可以通过mplayer播放一首音乐进行测试