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/*
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 */

/*
 * AIC3204 Tone
 *
 */
#include "stdio.h"
#include "usbstk5515.h"
extern Int16 AIC3204_rset( Uint16 regnum, Uint16 regval);

#define Rcv 0x08
#define Xmit 0x20

/* ----- *
 *
 * AIC3204 Loop
 * Output input from STEREO IN 1 through the HEADPHONE jack
 *
 * ----- */
Int16 aic3204_loop_stereo_in1( )
{
    /* Pre-generated sine wave data, 16-bit signed samples */
    Int16 j, i = 0;
    Int16 sample, data1, data2, data3, data4;

    /* Configure AIC3204 */
    AIC3204_rset( 0, 0);          // Select page 0
    AIC3204_rset( 1, 1);          // Reset codec
    AIC3204_rset( 0, 1);          // Point to page 1
    AIC3204_rset( 1, 8);          // Disable crude AVDD generation from DVDD
    AIC3204_rset( 2, 1);          // Enable Analog Blocks, use LDO power
    AIC3204_rset( 0, 0);          // Select page 0

    /* PLL and Clocks config and Power Up */
    AIC3204_rset( 27, 13);        // I2S, 16bits, BCLK e WCLK output, DOUT alta
    impedancia
    AIC3204_rset( 28, 0);        // Data ofset = 0
    AIC3204_rset( 4, 3);          // PLL setting: PLLCLK <- MCLK, CODEC_CLKIN <-PLLCLK
    AIC3204_rset( 6, 8);          // PLL setting: J=8
    AIC3204_rset( 7, 18);         // PLL setting: MSB D=4672
    AIC3204_rset( 8, 64);         // PLL setting: LSB D=4672
    AIC3204_rset( 29, 0);         // BDIV_CLKIN = DAC_CLK
    AIC3204_rset( 30, 132);       // BCLK / 4
    AIC3204_rset( 5, 145);        // PLL setting: Power up PLL, P=1 e R=1
    AIC3204_rset( 13, 0 );        // MSB for DOSR = 128
    AIC3204_rset( 14, 128);       // LSB for DOSR = 128
    AIC3204_rset( 20, 128);       // AOSR for AOSR = 128
    AIC3204_rset( 11, 134);       // Power up NDAC and set NDAC value to 6
    AIC3204_rset( 12, 134);       // Power up MDAC and set MDAC value to 6
    AIC3204_rset( 18, 134);       // Power up NADC and set NADC value to 6
    AIC3204_rset( 19, 134);       // Power up MADC and set MADC value to 6

    /* DAC ROUTING and Power Up */
    AIC3204_rset( 0, 1);          // Select page 1
    AIC3204_rset( 12, 8);         // LDAC AFIR routed to HPL
    AIC3204_rset( 13, 8);         // RDAC AFIR routed to HPR
    AIC3204_rset( 0, 0);          // Select page 0
    AIC3204_rset( 64, 2);         // Left vol=right vol
    AIC3204_rset( 65, 16);        // Left DAC gain to xdB VOL; Right tracks Left
    AIC3204_rset( 63, 0xd4);      // Power up left, right data paths and set channel
    AIC3204_rset( 0, 1);          // Select page 1
    AIC3204_rset( 16, 0x06);      // Unmute HPL , 6dB gain
    AIC3204_rset( 17, 0x06);      // Unmute HPR , 6dB gain
    AIC3204_rset( 9, 0x30);       // Power up HPL, HPR
    AIC3204_rset( 0, 0);          // Select page 0

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USBSTK5515_wait( 500 );    // Wait

/* ADC ROUTING and Power Up */
AIC3204_rset( 0, 1 );      // Select page 1
AIC3204_rset( 52, 0x30 );  // IN2_L to LADC_P through 40 kohm
AIC3204_rset( 55, 0x30 );  // IN2_R to RADC_P through 40 kohm
AIC3204_rset( 54, 0x03 );  // CM_1 (common mode) to LADC_M through 40 kohm
AIC3204_rset( 57, 0x03 );  // CM_1 (common mode) to RADC_M through 40 kohm
AIC3204_rset( 59, 0x00 );  // MIC_PGA_L unmute
AIC3204_rset( 60, 0x00 );  // MIC_PGA_R unmute
AIC3204_rset( 0, 0 );      // Select page 0
AIC3204_rset( 81, 0xc0 );  // Power up Left and Right ADC
AIC3204_rset( 82, 0x00 );  // Fine gain
AIC3204_rset( 83, 0x20 );  // Unmute Left ADC
AIC3204_rset( 84, 0x20 );  // Unmute Right ADC
AIC3204_rset( 0, 0 );
USBSTK5515_wait( 200 );    // Wait

/* I2S settings */
I2S0_SRGR = 0x0;
I2S0_CR = 0x8010;          // 16-bit word, slave, enable I2C
I2S0_ICMR = 0x3f;          // Enable interrupts

Config_DMA_I2S();

/* Play Tone */
//for ( i = 0 ; i < 5 ; i++ )
//{
    while (1)
    {
        //for ( j = 0 ; j < 1000 ; j++ )
        //{
            //for ( sample = 0 ; sample < 48 ; sample++ )
            //{
                /* Read Digital audio input */
                data3 = I2S0_W0_MSW_R; // 16 bit left channel received audio data
                data1 = I2S0_W0_LSW_R;
                data4 = I2S0_W1_MSW_R; // 16 bit right channel received audio data
                data2 = I2S0_W1_LSW_R;

                while((Rcv & I2S0_IR) == 0); // Wait for interrupt pending flag
                /* Write Digital audio input */
                I2S0_W0_MSW_W = data3; // 16 bit left channel transmit audio data
                I2S0_W0_LSW_W = 0;
                I2S0_W1_MSW_W = data4; // 16 bit right channel transmit audio data
                I2S0_W1_LSW_W = 0;
                while((Xmit & I2S0_IR) == 0); // Wait for interrupt pending flag
            }
        }
    }
//}
/* Disble I2S */
//I2S0_CR = 0x00;

return 0;
}

```