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#include <xs1.h>
#include <platform.h>
#include <stdio.h>
#include <syscall.h>
#include <xclib.h>
#include <debug_print.h>

#include "devicedefines.h"
#include "xc_ptr.h"
#include "audio.h"

/* audio I/O */
extern          in port      p_bitclk;           // MCLK = bit clock
extern buffered in port:32  p_frameclk;        //
extern buffered in port:32  p_tdm_in[3];       // 3 * I16S
extern buffered out port:32 p_tdm_out[3];      // 3 * I16S

/* audio clock */
extern clock          clk_audio_bclk;         // driven by
p_bitclk

//
int32_t audio_buf_0[NUM_USB_CHAN_OUT + NUM_USB_CHAN_IN]; //
outputs + inputs
int32_t audio_buf_1[NUM_USB_CHAN_OUT + NUM_USB_CHAN_IN]; //

/* 2nd decouple stage between usb core decouple and audio port
driver */
// void audio_decouple(chanend c_out, server interface if_dbl_buf
if_audio, unsigned device_config)
void audio_decouple(chanend c_out, chanend c_cfg, server interface
if_dbl_buf if_audio)
{
    unsigned sync_flag;
    unsigned sample;
    unsigned device_config;
    unsigned chan_count_out;
    unsigned chan_count_in;

    int32_t * movable mp_buffer = &audio_buf_0[0];

    // get config from Endpoint0()
    device_config = inuint(c_cfg);

    // debug_printf("device_config = %x\n", device_config);
    chan_count_out = (device_config >> 16) & 0xFF; //
NUM_USB_CHAN_OUT;
    chan_count_in = (device_config >> 24) & 0xFF; //
NUM_USB_CHAN_IN;

    //
    while(1)

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{
    select
    {
        case if_audio.swap(int32_t * movable &mp_new):

            /* swap movable pointers */
            int32_t * movable mp_tmp = move(mp_new);
            mp_new = move(mp_buffer);
            mp_buffer = move(mp_tmp);
            /* set sync flag to trigger signal processing
*/
            sync_flag = 1;

            break;

        default:
            if(sync_flag)
            {
                /* sync with decouple() - command =
deliver(c_mix_out); */
                outuint(c_out, 0);

                #pragma loop unroll
                for(unsigned i = 0; i < NUM_USB_CHAN_OUT;
i++)
                {
                    // mp_buffer[i] =
bitrev(inuint(c_out));
                    sample = bitrev(inuint(c_out));
                    if(i < chan_count_out)
                        mp_buffer[i] = sample;
                    else
                        mp_buffer[i] = 0;
                }

                #pragma loop unroll
                for(unsigned i = 0; i < NUM_USB_CHAN_IN; i+
+)
                {
                    // outuint(c_out,
bitrev(mp_buffer[NUM_USB_CHAN_OUT + i]));
                    #if USB_LOOPBACK
                        sample = bitrev(mp_buffer[i]);
                    #else
                        sample =
bitrev(mp_buffer[NUM_USB_CHAN_OUT + i]);
                    #endif
                    if(i < chan_count_in)
                        outuint(c_out, sample);
                    else
                        outuint(c_out, 0);
                }

                /* reset sync flag */

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        sync_flag = 0;
    }
    break;
}
}

//
#define SYNC_COMPARE    0x7FFFFFFF

//
unsigned i16s_in_buf [48 << 1];    // double buffer
unsigned i16s_out_buf[48 << 1];    //
unsigned resync_cnt = 0;

//
void audio_port_driver(client interface if_dbl_buf if_audio)
{
    unsigned sync_errors = 0;
    unsigned resync_flag;
    unsigned fsync_data;
    unsigned fsync_buf[16];
    unsigned timeslot;
    unsigned portclk;
    unsigned tmp;

    unsigned adc_tmp[48];

    // initialize movable pointer (for double buffering)
    int32_t * movable mp_audio = &audio_buf_1[0];

    //
    // debug_printf("start I16S transceiver ... \n");

    //
    set_thread_fast_mode_on();

    // configure_audio_ports_for_streaming();
    // wait until FSYNC is low
    tmp = 1;
    while (tmp)
        p_frameclk := tmp;

    stop_clock(clk_audio_bclk);

    clearbuf(p_tdm_out[0]);
    clearbuf(p_tdm_out[1]);
    clearbuf(p_tdm_out[2]);
    clearbuf(p_tdm_in[0]);
    clearbuf(p_tdm_in[1]);
    clearbuf(p_tdm_in[2]);
    clearbuf(p_frameclk);

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set_port_inv(p_bitclk);
set_pad_delay(p_bitclk, 5);    // 10 ns
set_port_no_inv(p_frameclk);  //

// generate bit clock block from pin - TDM: bit clock = master
clock
configure_clock_src(clk_audio_bclk, p_bitclk);
// clock output data ports from clock block
// output ports are shifted at falling (internal) clock edges
configure_out_port_no_ready(p_tdm_out[0], clk_audio_bclk, 0);
configure_out_port_no_ready(p_tdm_out[1], clk_audio_bclk, 0);
configure_out_port_no_ready(p_tdm_out[2], clk_audio_bclk, 0);
// clock input data ports from clock block
// inputs are sampled at rising (internal) clock edges
configure_in_port_no_ready(p_tdm_in[0], clk_audio_bclk);
configure_in_port_no_ready(p_tdm_in[1], clk_audio_bclk);
configure_in_port_no_ready(p_tdm_in[2], clk_audio_bclk);
configure_in_port_no_ready(p_frameclk, clk_audio_bclk);
// shift sampling of input ports to falling (internal) clock
edges
set_port_sample_delay(p_tdm_in[0]);
set_port_sample_delay(p_tdm_in[1]);
set_port_sample_delay(p_tdm_in[2]);
set_port_sample_delay(p_frameclk);

start_clock(clk_audio_bclk);

//
resync_flag = 1;

// wait for audio clocks stable
fsync_data = 0;
for (int i = 0; i < 50; i++)
{
    while (!fsync_data)    // wait for rising FSYNC edge
        p_frameclk := fsync_data;
    while (fsync_data)    // wait for falling FSYNC edge
        p_frameclk := fsync_data;
}

// audio driver loop ( init/reinit - loop)
while (1)
{
    // init
    if (resync_flag)
    {
        resync_flag = 0;
        sync_errors = 0;
        resync_cnt ++;
    }
}

#if 0
    // wait until MCLK is low
    tmp = 1;
#endif

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while (tmp)
    p_mclk := tmp;
#endif

stop_clock(clk_audio_bclk);

clearbuf(p_tdm_out[0]);
clearbuf(p_tdm_out[1]);
clearbuf(p_tdm_out[2]);
clearbuf(p_tdm_in[0]);
clearbuf(p_tdm_in[1]);
clearbuf(p_tdm_in[2]);
clearbuf(p_frameclk);

start_clock(clk_audio_bclk);

/*
// wait for fsync = HIGH
fsync_data = 0;
while (fsync_data != 0xffffffff)
    p_frameclk := fsync_data;

// catch falling FSYNC edge
while (fsync_data == 0xffffffff)
    p_frameclk := fsync_data @ portclk;

// calculate delay to start data streaming in sync to
frame start (falling edge of FSYNC)
delay_in = portclk + 4096; // has to be a
multiple of 512 to match both formats (I8S / I16S)

// 48 KHz
delay_in += 448 - 1; // align to 2 samples
before frame sync

while (fsync_data != 0x7FFFFFFF)
{
    delay_in--;
    fsync_data <<= 1; // fsync_data is
bit-reversed...
    fsync_data |= 1;
}

delay_out = delay_in + 32 + 1;

// timed in sample 0 -2
p_tdm_in[0] @ (delay_in) := void; // synced dummy
inputs
p_tdm_in[1] @ (delay_in) := void;
p_tdm_in[2] @ (delay_in) := void;
p_frameclk @ (delay_in) := fsync_data;

// timed out
p_tdm_out[0] @ (delay_out) <: 0xffffffff;

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        p_tdm_out[1] @ (delay_out) <: 0xffffffff;
        p_tdm_out[2] @ (delay_out) <: 0xffffffff;

        p_tdm_in[0] :=> void;                // synced dummy
inputs
        p_tdm_in[1] :=> void;
        p_tdm_in[2] :=> void;
        p_frameclk :=> void;
*/

        p_frameclk when pinseq(1) :=> void;
        p_frameclk when pinseq(0) :=> void @ portclk;

        unsigned initial_out_port_time = portclk - 32 + (16*32);
        unsigned initial_in_port_time  = portclk - 32 +
(16*32)+32 - 1;

        p_tdm_out[0] @ initial_out_port_time <: mp_audio[15];
        p_tdm_out[1] @ initial_out_port_time <: mp_audio[31];
        p_tdm_out[2] @ initial_out_port_time <: mp_audio[47];

        // XC doesn't have syntax for setting a timed input
without waiting for the input
        asm("setpt res[%0],
%1"::"r"(p_frameclk),"r"(initial_in_port_time));

        asm("setpt res[%0],
%1"::"r"(p_tdm_in[0]),"r"(initial_in_port_time));
        asm("setpt res[%0],
%1"::"r"(p_tdm_in[1]),"r"(initial_in_port_time));
        asm("setpt res[%0],
%1"::"r"(p_tdm_in[2]),"r"(initial_in_port_time));
    }

    /* 48 KHz / 48 ch - I16S */
    for (timeslot = 0; timeslot < 16; timeslot++)
    {

#ifdef AUDIO_LOOPBACK
        p_tdm_out[0] <: mp_audio[NUM_USB_CHAN_OUT + timeslot +
0];
        p_tdm_out[1] <: mp_audio[NUM_USB_CHAN_OUT + timeslot +
16];
        p_tdm_out[2] <: mp_audio[NUM_USB_CHAN_OUT + timeslot +
32];
#else
        p_tdm_out[0] <: mp_audio[timeslot + 0];
        p_tdm_out[1] <: mp_audio[timeslot + 16];
        p_tdm_out[2] <: mp_audio[timeslot + 32];
#endif

        p_frameclk :=> fsync_buf[timeslot]; // fsync_data;

        if (timeslot == 0)

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        {
            p_tdm_in[0] := mp_audio[NUM_USB_CHAN_OUT + timeslot
+ 15];
            p_tdm_in[1] := mp_audio[NUM_USB_CHAN_OUT + timeslot
+ 31];
            p_tdm_in[2] := mp_audio[NUM_USB_CHAN_OUT + timeslot
+ 47];
        }
        else
        {
            p_tdm_in[0] := adc_tmp[timeslot - 1];
            p_tdm_in[1] := adc_tmp[timeslot + 15];
            p_tdm_in[2] := adc_tmp[timeslot + 31];
        }

        //
        if (timeslot < 15) {
            /* fill audio double buffer with buffered adc values
*/
            mp_audio[NUM_USB_CHAN_OUT + timeslot + 0] =
adc_tmp[timeslot + 0];
            mp_audio[NUM_USB_CHAN_OUT + timeslot + 16] =
adc_tmp[timeslot + 16];
            mp_audio[NUM_USB_CHAN_OUT + timeslot + 32] =
adc_tmp[timeslot + 32];
        }
    }

    // make sure serial ports are still in sync with FSYNC
    if (fsync_buf[0] != 0x7FFFFFFF)
    {
        // wrong alignment to FSYNC
        sync_errors++;
        if (sync_errors > 5)
        {
            resync_flag = 1;
            debug_printf("resync = %d\n", resync_cnt);
        }
    }

    /* swap audio buffer pointers (requires about 300 ns) */
    if_audio.swap(mp_audio);

} // while(1)
}

```