

Audio spectrum analyzer

Project Number: AR1731

Abstract

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Digital Signal Processing has become very important part of modern electronics. Although the best way to perform digital signal processing is to use a special DSP processor, it is possible to achieve good result when using fast conventional processor or microcontroller. This article describes how to build simple audio spectrum analyzer based on FFT algorithm using Philips LPC2138 microcontroller.

Fig.1 shows block diagram of spectrum analyzer, **fig.2** contains detailed schematic of ZL6ARM prototype board which has been used to build spectrum analyzer. ZL6ARM board contains LPC2138 microcontroller.

Input audio signal is sampled by LPC2138 microcontroller which uses built-in analog to digital converter. Sampling frequency equals 40kHz. After getting 256 samples of the signal the 256-point FFT (*Fast Fourier Transform*) algorithm is performed. The result of FFT consists of 256 complex numbers. Real and imaginary parts of those numbers are divided by 128 and absolute values of them are displayed. They are amplitude spectrum of the input signal.

Displaying the spectrum using simple graphic LCD displays is not a good idea due to relatively bad resolution of these displays. That's why I decided that spectrum will be send to PC computer via RS232 interface and there displayed by special created Windows application. It has to be stressed, that PC is used only for displaying the spectrum and all the job is made by LPC2138.

There has been designed special complex type (named *cplx*) to represent complex numbers. It's common knowledge that a complex number is a vector composed of two real numbers. Representing real (fractional) numbers was one of the main issues. Due to lack of FPU (*Floating Point Unit*) in ARM architecture the fixed point numbers with 32-bit width mantissa have been used to represent fractional numbers. Another issue was of course an implementation of FFT algorithm. Spectrum analyzer allows to use one of three window functions: rectangular window, Hanning window and Hamming window. These three windows have been chosen because they are the most popular windows used in DSP. Any other window function can be added by simple source code modification.

List.1 shows main loop of spectrum analyzer software. The microcontroller performs following actions periodically:

1. Getting 256 samples of input signal;
2. Multiplying samples by window function chosen by the user;
3. Computing 256-point FFT;

4. Computing squares of absolute values of FFT result (squares of amplitude spectrum);
5. Checking if any request came from PC;
6. Sending spectrum when request received.

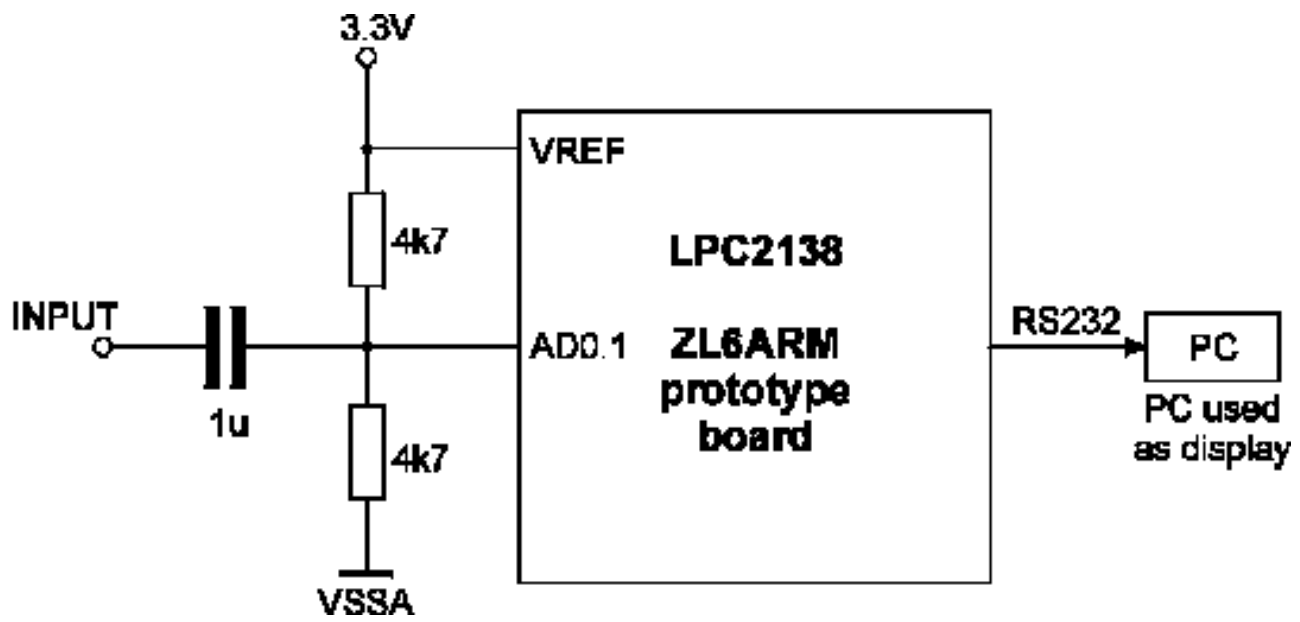
List.2 shows implementation of fixed point arithmetic. The main advantage of fixed point numbers is that arithmetic operations on fractional numbers can be done using only operations on integer numbers. The main disadvantage is risk of overflow while multiplying or dividing. The discussion about this risk is one of article's issues. **List.3** shows implementation of complex number type *cplx* which is vector of two real numbers (*fix*). **List.4** shows a heart of the project's code – implementation of FFT algorithm using prior designed *cplx* and *fix* data types.

Fig.3 contains screenshot of PC application window made while displaying spectrum of square wave of 1kHz frequency, 0.75V amplitude and 50% duty factor. Rectangular window has been used and big DFT leakage is visible. Using Hanning window (**fig.4**) significantly decreases the leakage. Discussion about cause of leakage is also one of article's issues.

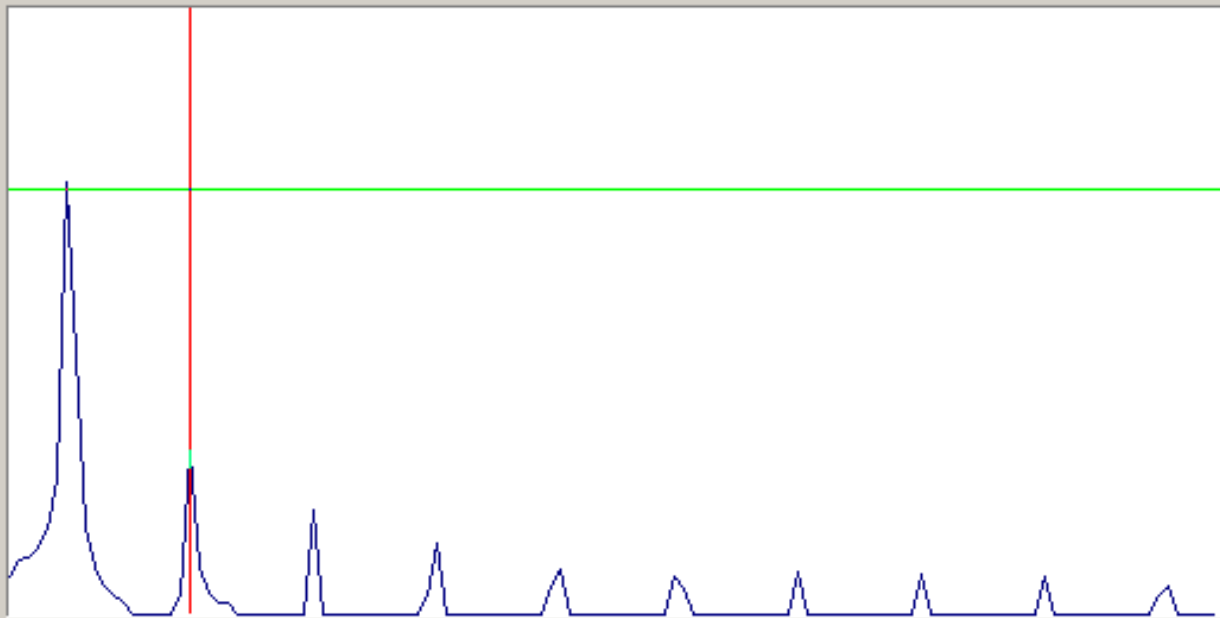
[1] Richard G. Lyons “Understanding Digital Signal Processing”

[2] “Fixed Point Arithmetic on the ARM” ARM DAI 0033A, Advanced RISC Machines Ltd (ARM) 1996 [pdf]

[3] “LPC2131/2132/2138 User Manual Preliminary Release”, Philips Semiconductors 2004 [pdf]



ARM Spectrum Analyzer - Entry: AR1731



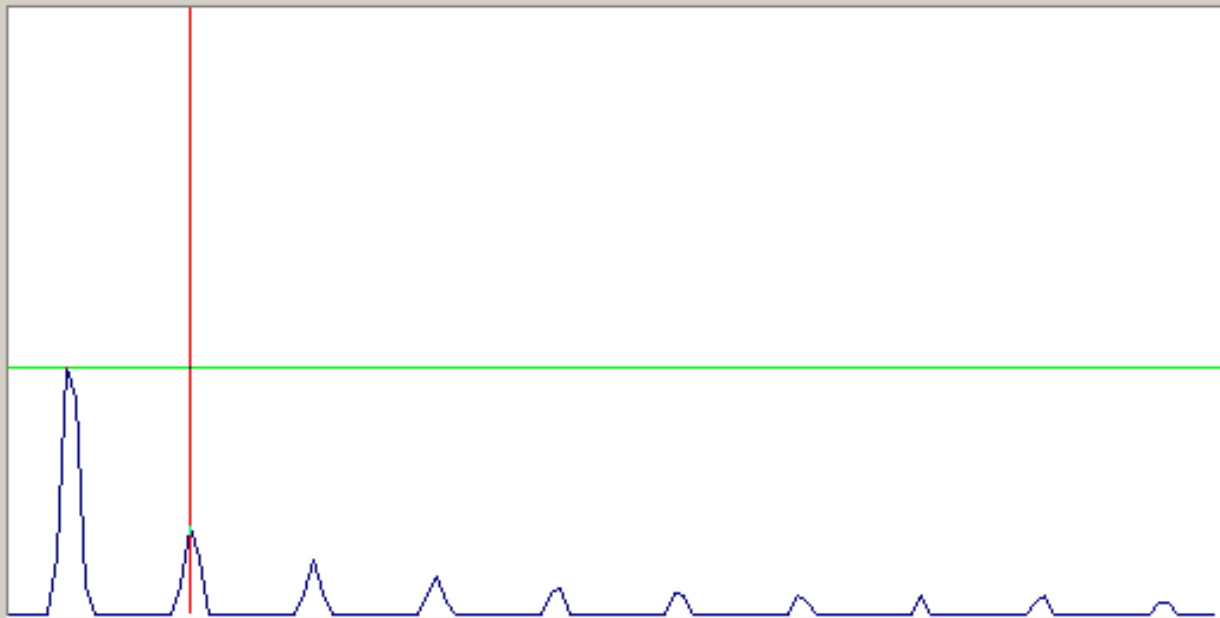
X: Y: F: 2,97kHz V: 0,70V

B L

Run

Window: None

ARM Spectrum Analyzer - Entry: AR1731



X: Y: F: 2,97kHz V: 0,41V

B L

Run

Window: Hanning



List. 1

```

while(1)
{
    unsigned long i;

    //Sampling
    Sampling(Samples);

    //Multiplying by window's factors
    if(WindowType==wtHanning)
        Hanning(Samples);
    else if(WindowType==wtHamming)
        Hamming(Samples);

    //Computing Spectrum (FFT)
    FFT(Samples, FFTResult);

    //Computing squared modules of FFTResult (Spectrum)
    for(i=0; i<N; i++)
    {
        fix Re2, Im2;

        //Scaling by N/2=128
        FFTResult[i].Re>>=7;
        FFTResult[i].Im>>=7;

        //Squaring
        Re2=FMUL(FFTResult[i].Re, FFTResult[i].Re, q);
        Im2=FMUL(FFTResult[i].Im, FFTResult[i].Im, q);
        Spectrum[i]=Re2+Im2;
    }

    //Changing the window
    if(!SW1 | !SW2 | !SW3)
    {
        delay(0x800);
        if(!SW1)
            WindowType=wtNone;
        else if(!SW2)
            WindowType=wtHanning;
        else if(!SW3)
            WindowType=wtHamming;

        DisplayWindowType(WindowType);
    }

    //Checking request
    if((UOLSR & 0x01)==0)
        continue;

    //Sending spectrum to PC (if requested)
    SendSpectrumInfo(Spectrum, WindowType);

    //Empty input buffer by dummy reading
    while(UOLSR & 0x01)
        i=UORBR;
}

```

List. 2

```
#ifndef FIX_H
#define FIX_H

extern const int q; //exponent

//fixed point number type
typedef int fix; //mantissa (interpreted as fixed point number with exponent
q)

//arithmetic operations
#define FADD(a, b) ((a)+(b))
#define FSUB(a, b) ((a)-(b))
#define FMUL(a, b, q) (((a)*(b))>>(q))
#define FDIV(a, b, q) (((a)<<(q))/(b))

#endif
```

List. 3

```
typedef struct
{
    fi x Re;
    fi x Im;
} cpl x;
```

List. 4

```

//*****
// Butterfly function
// Input, output: x, y
// x <- x + Wn[k]*y
// y <- x - Wn[k]*y
//*****
void Butterfly(cplx *x, cplx *y, int k)
{
    cplx tx, ty, tWy;

    tWy=cplx_mul(Wn[k], *y);
    tx=cplx_add(*x, tWy);
    ty=cplx_sub(*x, tWy);

    *x=tx;
    *y=ty;
}

//*****
// FFT function
// x - input N complex values
// X - output N complex values
//*****
void FFT(cplx *x, cplx *X)
{
    int stc;          //stages counter
    int blc;          //blocks of butterflies in stage counter
    int bfc;          //butterflies in block counter

    int NumOfStages;    //Number of stages = log2(N)
    int NumOfBlocks;    //in stage
    int NumOfButterflies; //in block

    //Determining number of stages - counting log2(N)
    for(NumOfStages=0; NumOfStages<32; NumOfStages++)
        if(N==(1<<NumOfStages))
            break;

    //Copying input data to output buffer
    //to perform in-place computing
    for(stc=0; stc<N; stc++)
        X[stc]=x[BiThift(stc, NumOfStages-1)];

    //Main FFT loop
    for(stc=0; stc<NumOfStages; stc++)
    {
        NumOfButterflies=(1<<stc);
        NumOfBlocks=N>>(stc+1);

        for(blc=0; blc<NumOfBlocks; blc++)
        {
            int base=(1<<(stc+1))*blc;
            for(bfc=0; bfc<NumOfButterflies; bfc++)
                Butterfly(X+base+bfc, X+base+bfc+NumOfButterflies, NumOfBlocks*bfc);
        }
    }
}

```