DSP 2nd Project — DTMF Detection

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See my Github Repository for more infomation: $\label{eq:https://github.com/lzhbrian/DTMF} https://github.com/lzhbrian/DTMF$

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1 Problem

要求利用 FFT, Goertzel 算法,对给定音频文件中的双音多频信号进行检测和识别。

- 1. 下载附件包中第一小题的 10 个长度不一的音频文件,利用第一次课程设计中编写的 FFT 程序对这 10 个文件中的 DTMF 信号进行频谱分析,最后给出 10 个文件所对应的真实数字。
- 2. 编写 Goertzel 算法的 C/C++ 语言程序,完成(1)中的要求。
- 3. 下载附件包中第二小题的一个长音频文件,文件中包含了一串 DTMF 信号,每个双音多频信号之间的时间间隔不一,对本串 DTMF 信号进行识别。

2 Solution

2.1 Read the .wav files — read_wav.m

We use a Matlab audioread() function to convert the .wav file to .txt file, extracting the time-zone signals. Saving .txt files to ./txtData1 and ./txtData2.

The code is shown in section 3.8

2.2 Realization of DTMF using FFT — dif_fft.h

This part is explained in the previous report. The code is shown in section 3.5 See https://github.com/lzhbrian/Fast-Fourier-Transform for more information.

2.3 Realization of DTMF using Goertzel — goertzel.h

Goertzel algorithm is a method by which we can only calculate the amplitude of certain frequency. By the following functions, We can obtain the X[k] we want:

$$v_k[n] = x[n] + 2\cos(\omega_k)v_k[n-1] - v_k[n-2] \tag{1}$$

$$X[k] = v_k[N-1] - W_N^k v_k[N-2]$$
(2)

where

$$\omega_k = 2\pi k/N, W_N = e^{2\pi/N}, v_k[-2] = v_k[-1] = 0, v_k[0] = x[0]$$

In this DTMF detection, we want to acquire the amplitude of

$$697Hz, 770Hz, 852Hz, 941Hz, 1209Hz, 1336Hz, 1477Hz, 1633Hz, 1477Hz, 1633Hz, 1477Hz, 1633Hz, 1477Hz, 1633Hz, 1477Hz, 1633Hz, 1634Hz, 1634Hz, 1634Hz, 1634Hz, 1634Hz, 1634Hz, 1634Hz, 1634Hz, 1634Hz, 1644Hz, 1644Hz,$$

Note that we get the k for each frequency by the following equation:

$$k = (N * f)/SamplingRate;$$
 (3)

where N is the length of the sequence, and f is the targeted frequency.

So we first iteratively calculate the value of $v_k[N-1]$ and $v_k[N-2]$ using equation (1), then we use them to get the value of X[k] by equation (2). In the real practice, we further return the amplitude of X[k] by calculating their sum of squares.

The code is shown in section 3.6

2.4 DTMF Detection — find_dtmf_symbol.h

For convenience, we wrote a function to obtain the symbol of a DTMF signal. By inputing the max two frequency, we can get the symbol 1, 2, 3, 4, 5, 6, 7, 8, 9, A, B, C, D, 0, *, #

The code is shown in section 3.7

2.5 Recognition of Dataset 1 — 10 signals

2.5.1 FFT — DTMF_1.cpp

The result is

5, 1, 6, 9, 8, 7, 3, 4, 0, 2

for data

1081, 1107, 1140, 1219, 1234, 1489, 1507, 1611, 1942, 1944

respectively, as shown in Table 1. The code is shown in section 3.1.

Signal length	1st freq	2nd freq	symbol
1081	769	1337	5
1107	696	1212	1
1140	770	1477	6
1219	852	1477	9
1234	852	1337	8
1489	852	1212	7
1507	696	1477	3
1611	770	1212	4
1942	942	1337	0
1944	696	1337	2

Table 1: DTMF result for 10 respective signals

2.5.2 Goertzel — DTMF_2.cpp

Goertzel has shown exactly same result with FFT. The code is shown in section 3.2.

2.6 Recognition of Dataset 2 — A long signal, DTMF_3.cpp

2.6.1 Judge the start-end time

By Matlab, we can obtain the time-zone signals as shown in Figure 1. We then manually get the start, end time of each of signal as shown in Table 2.

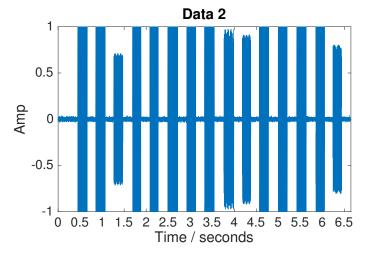


Figure 1: Timezone signal of Data 2

#	Start time	End time	#	Start time	End time
1	0.45	0.66	9	3.78	4.00
2	0.86	1.07	10	4.18	4.40
3	1.27	1.47	11	4.58	4.79
4	1.70	1.89	12	5.01	5.22
5	2.09	2.29	13	5.43	5.64
6	2.50	2.72	14	5.87	6.06
7	2.92	3.10	15	6.26	6.45
8	3.34	3.55			

Table 2: Start-End time of each signal

2.6.2 Result

By the Goertzel method, we can obtain the symbols for the long signal are:

$$2, 0, 5, 8, 9, 1, 1, 3, 2, 0, u, 4, 6, 4, 9$$

u represents unknown, i.e. we cannot detect the 11th symbol, its amplitude response is shown in Figure 2.From the figure, we can see that we don't have two target frequencies which have a significant amplitude, s.t. we cannot decipher this symbol. The code is shown in section 3.3.

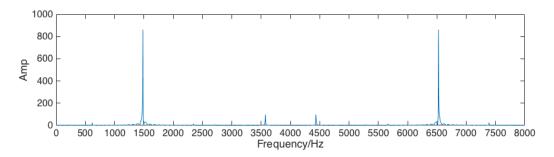


Figure 2: Amplitude response of the 11th symbol

3 Code

3.1 DTMF_1.cpp — Main function for problem 1

```
// Work by Lin, Tzu-Heng
// W42, 2014011054
// Dept. of Electronic Engineering, Tsinghua University
// DSP Course Work
# include <fstream>
# include <string>
# include <stdio.h>
# include <stdlib.h>
# include <cmath>
# include <ctime>
# include <sstream>
# include <iostream>
# include <cstdlib>
# include <unistd.h>
# include <dirent.h>
# include <sys/stat.h>
# define PI 3.1415926
int const MAX_STR_LEN = 200;
# include "complex.h"
# include "dif_fft.h"
# include "find_dtmf_symbol.h"
using namespace std;
// FFT Algorithm
int main()
    \begin{array}{ll} char & dir\_name [100] \ = \ "./txtData1/"; \end{array}
    struct dirent * filename;
                                   // return value for readdir()
    DIR * dir;
                                   // return value for opendir()
    dir = opendir( dir_name );
    /* read all the files in the dir \sim */
    while( ( filename = readdir(dir) ) != NULL )
        // get rid of "." and ".."
        if (strcmp (filename->d_name, ".") == 0 ||
            strcmp(filename -> d_name, "..") == 0)
             continue;
        cout << filename ->d_name <<endl;</pre>
        char hhh[MAX_STR_LEN];
        strcpy(hhh, dir_name);
        strcat(hhh, (filename->d_name));
        ifstream in ( hhh );
```

```
string filename;
string line;
if(in) // 有该文件
    double audio [60000];
    int count = 0;
    while (getline (in, line)) // line中不包括每行的换行符
        // audio[count] = atof(const_cast<const_char *>(line.c_str()));
        double d;
        stringstream ss(line);
        ss \gg d;
        audio [count] = d;
        count = count + 1;
        // cout << audio [count -1] << endl;
   }
   // is 2<sup>k</sup> or not, if not add zero
    int add_zero_count = 0;
    for (int i = 1; i < 100; ++i)
    {
        if (pow(2,i) < count) {
            continue;
        else if (pow(2,i) = count) {
            break;
        } else {
            add\_zero\_count = pow(2,i) - count;
            break;
        }
    }
   // add value to input_seq[]
    int total_length = count + add_zero_count;
   complex* input_seq = new complex[total_length];
    for (int i = 0; i < count; i++)
        input_seq[i].re = audio[i];
        input\_seq[i].im = 0;
    }
    for (int i = 0; i < add_zero_count; i++)
        input\_seq[count + i].re = 0;
       input\_seq[count + i].im = 0;
    }
   // FFT
   complex* output_seq = DIF_FFT_reordered(input_seq, total_length);
   // Amp, Find max 1,2 and their positions
```

int amp;

```
int max1 = 0;
             int max2 = 0;
             int max1_pos = 0;
             int max2_pos = 0;
             for (int i = 0; i < total_length/2; ++i)
                 amp = pow(output_seq[i].re,2) + pow(output_seq[i].im,2);
                 if (amp > max2)
                     if (amp > max1) {
                         // \max 2 = \max 1
                         \max 2 = \max 1;
                         \max 2 pos = \max 1 pos;
                          // \max 1 = \text{new}
                         \max 1 = amp;
                         \max 1 \_pos = i;
                     } else {
                         \max 2 = \min;
                         \max 2 \_pos = i;
                     }
                 }
             }
             // x-axis
                 // 0: f_s/(N-1): f_s
             int fs = 8000;
             int N = total_length;
             double step = double(fs)/(N-1);
             double* x_axis = new double[N];
             for (int i = 0; i < N; ++i)
                 x_axis[i] = i * step;
             // Get max1, max2 freq
             cout << "Max1_pos:" << x_axis[max1_pos] << endl;
             cout \ll "Max2_pos:" \ll x_axis[max2_pos] \ll endl;
             // decipher
             char output_symbol = find_dtmf_symbol(x_axis[max1_pos], x_axis[max2_pos
             cout << "The_symbol_for_this_sound:_" << output_symbol << endl;
             cout << endl;
         } else { // fail reading file
             cout << "nousuchufile" << endl;
    return 0;
}
```

3.2 DTMF_2.cpp — Main function for problem 2

```
// Work by Lin, Tzu-Heng
// W42, 2014011054
// Dept. of Electronic Engineering, Tsinghua University
// DSP Course Work
# include <fstream>
# include <string>
# include <stdio.h>
# include <stdlib.h>
# include <cmath>
# include <ctime>
# include <sstream>
# include <iostream>
# include <cstdlib>
# include <unistd.h>
# include <dirent.h>
# include <sys/stat.h>
# define PI 3.1415926
int const MAX_STR_LEN = 200;
# include "complex.h"
# include "goertzel.h"
# include "find_dtmf_symbol.h"
using namespace std;
// Goertzel Algorithm
int main()
    cout << "Running_Prob2,_using_Goertzel_Algorithm_..._" << endl << endl;
    char dir_name[100] = "./txtData1/";
    struct dirent * filename;
                                 // return value for readdir()
    DIR * dir;
                                  // return value for opendir()
    dir = opendir( dir_name );
    /* read all the files in the dir \sim */
    while( ( filename = readdir(dir) ) != NULL )
        // get rid of "." and ".."
        if (strcmp (filename->d_name, ".") == 0 ||
            strcmp(filename->d_name, "..") == 0)
            continue;
        cout << filename ->d_name <<endl;</pre>
        char hhh [MAX_STR_LEN];
        strcpy(hhh, dir_name);
```

```
strcat(hhh, (filename->d_name));
ifstream in ( hhh );
string filename;
string line;
if(in) // 有该文件
    double audio [60000];
    int count = 0;
    while (getline (in, line)) // line中不包括每行的换行符
        // audio [count] = atof(const_cast<const_char *>(line.c_str()));
        double d;
        stringstream ss(line);
        ss \gg d;
        audio [count] = d;
        count = count + 1;
        // cout << audio [count - 1] << endl;
    }
    // is 2<sup>k</sup> or not, if not add zero
    int add_zero_count = 0;
    for (int i = 1; i < 100; ++i)
        if (pow(2,i) < count) {
            continue;
        else if (pow(2,i) = count) {
            break;
        } else {
            add_zero_count = pow(2, i) - count;
            break;
        }
    }
    // add value to input_seq[]
    int total_length = count + add_zero_count;
    complex* input_seq = new complex[total_length];
    for (int i = 0; i < count; i++)
    {
        input_seq[i].re = audio[i];
        input\_seq[i].im = 0;
    for (int i = 0; i < add\_zero\_count; i++)
    {
        input\_seq[count + i].re = 0;
        input\_seq[count + i].im = 0;
    }
    // Goertzel, return amp^2
    double* targeted_amp = Goertzel(input_seq, total_length);
```

```
// Amp, Find max 1,2 and their positions
        int amp;
        int max1 = 0;
        int max2 = 0;
        int max1\_pos = 0;
        int max2_pos = 0;
        for (int i = 0; i < 8; ++i)
             amp = targeted_amp[i];
             if (amp > max2)
                 if (amp > max1) {
                      // \max 2 = \max 1
                      \max 2 = \max 1;
                      \max 2 \_ pos = \max 1 \_ pos;
                      // \max 1 = \text{new}
                      \max 1 = amp;
                      \max 1 \_pos = i;
                 } else {
                      \max 2 = \min;
                      \max 2 \_ pos = i;
                 }
             }
        }
        // Get max1, max2 freq
        double x_axis[] =
                                                     \{697,
                                                      770,
                                                      852,
                                                      941,
                               1209,1336,1477,1633};
        cout \ll "Max1_pos:" \ll x_axis[max1_pos] \ll endl;
        cout << "Max2\_pos:" << x\_axis[max2\_pos] << endl;
        // decipher
        char output_symbol = find_dtmf_symbol(x_axis[max1_pos], x_axis[max2_pos
        cout << "The_symbol_for_this_sound:_" << output_symbol << endl;
        cout <\!\!< endl;
    } else { // fail reading file
        cout << "nousuchufile" << endl;
return 0;
```

3.3 DTMF_3.cpp — Main function for problem 3

```
// Work by Lin, Tzu-Heng
// W42, 2014011054
// Dept. of Electronic Engineering, Tsinghua University
// DSP Course Work
# include <fstream>
# include <string>
# include <stdio.h>
# include <stdlib.h>
# include <cmath>
# include <ctime>
# include <sstream>
# include <iostream>
# include <cstdlib>
# include <unistd.h>
# include <dirent.h>
# include <sys/stat.h>
# define PI 3.1415926
int const MAX_STR_LEN = 200;
# include "complex.h"
# include "goertzel.h"
# include "find_dtmf_symbol.h"
using namespace std;
// Goertzel Algorithm to identify 3
int main()
{
         cout << "Identifying \sqcup Prob3 , \sqcup a \sqcup long \sqcup audio , \sqcup using \sqcup Goertzel \sqcup Algorithm \sqcup \dots " <<
             endl << endl;
         ifstream in("./txtData2/data.txt");
         string filename;
         string line;
         if(in) // 有该文件
             double audio [60000];
             int count = 0;
             while (getline (in, line)) // line中不包括每行的换行符
                 // audio[count] = atof(const_cast<const_char *>(line.c_str()));
                 double d;
                 stringstream ss(line);
                 ss \gg d;
                 audio [count] = d;
```

```
count = count + 1;
    // cout << audio [count -1] << endl;
}
// Stops (in seconds)
double stops[] =
                     \{0.4459, 0.6633,
                      0.8642, 1.072,
                      1.27,
                              1.473,
                      1.7,
                              1.89,
                      2.09,
                              2.29,
                              2.722,
                      2.5,
                      2.915, 3.1,
                      3.34,
                              3.549,
                      3.782, 4,
                      4.182,
                              4.398,
                      4.581, \quad 4.794,
                      5.013, 5.222,
                      5.431, 5.641,
                      5.869, 6.059,
                      6.258, 6.447;
int len_data2 = count;
double seconds_len_data2 = len_data2/8000.0;
for (int d = 0; d < 15; ++d)
{
    int start_index = stops[d*2]/seconds_len_data2*len_data2;
    int end_index = stops[d*2+1]/seconds_len_data2*len_data2;
    // Get this audio
    count = end_index - start_index + 1;
    double* this_audio = new double[count];
    for (int i = 0; i < count; ++i)
        this_audio[i] = audio[i+start_index];
    }
    // is 2<sup>k</sup> or not, if not add zero
    int add_zero_count = 0;
    for (int i = 1; i < 100; ++i)
        if (pow(2,i) < count) {
            continue;
        } else if ( pow(2,i) = count ) {
            break;
        } else {
            add\_zero\_count = pow(2,i) - count;
            break;
        }
    }
```

```
// add value to input_seq[]
int total_length = count + add_zero_count;
complex* input_seq = new complex[total_length];
for (int i = 0; i < count; i++)
    input_seq[i].re = this_audio[i];
    input\_seq[i].im = 0;
for (int i = 0; i < add_zero_count; i++)
    input\_seq[count + i].re = 0;
    input\_seq[count + i].im = 0;
}
// Goertzel, return amp^2
double* targeted_amp = Goertzel(input_seq, total_length);
// amp indicator
// for (int ii = 0; ii < 8; ++ii)
// {
// cout << targeted_amp[ii] << endl;
// }
// Amp, Find max 1,2 and their positions
int amp;
int max1 = 0;
int max2 = 0;
int max1_pos = 0;
int max2_pos = 0;
for (int i = 0; i < 8; ++i)
    amp = targeted_amp[i];
    if (amp > max2)
    {
         if (amp > max1) {
             // \max 2 = \max 1
             \max 2 = \max 1;
             \max 2 pos = \max 1 pos;
             // \max 1 = \text{new}
             \max 1 = amp;
             \max 1\_pos = i;
        } else {
             \max 2 = \sup;
             \max 2 \_pos = i;
    }
}
// Get max1, max2 freq
double x_axis[] =
                                           {697,
                                            770,
```

```
852,
941,
1209,1336,1477,1633};

cout << "Maxl_pos:" << x_axis[maxl_pos] << endl;
cout << "Max2_pos:" << x_axis[max2_pos] << endl;

// decipher
char output_symbol = find_dtmf_symbol(x_axis[max1_pos], x_axis[max2_pos]);
cout << "The_symbol_for_this_sound:_" << output_symbol << endl;

cout << endl;

}

else { // fail reading file
cout << "no_such_file" << endl;
}

return 0;
}
```

3.4 complex.h

```
// Work by Lin, Tzu-Heng
// W42, 2014011054
// Dept. of Electronic Engineering, Tsinghua University
// DSP Course Work
   ************************************
// Complex Struct & Some Basic func
using namespace std;
typedef struct Complex
   double re;
   double im;
   Complex() {
       re = 0;
       im = 0;
   };
   Complex (double a, double b) {
       re = a;
       im = b;
   };
} complex;
```

```
complex*\ append\_seq(complex\ seq\_1\,[]\ ,\ complex\ seq\_2\,[]\ ,\ int\ N)\,;
complex* reorder_seq(complex input_seq[], int N);
complex* Calc_WN(int N);
int reverse_bit(int value, int N);
// Multiplier
complex ComplexMul(complex c1, complex c2)
    complex r;
    r.re = c1.re*c2.re - c1.im*c2.im;
    r.im = c1.re*c2.im + c1.im*c2.re;
    return r;
}
// Adder
complex Complex Add(complex c1, complex c2)
    complex r;
    r.re = c1.re + c2.re;
    r.im = c1.im + c2.im;
    return r;
}
complex ReverseComplex(complex c)
    c.re = -c.re;
    c.im = -c.im;
    return c;
}
// scalar mul
complex ComplexScalarMul(complex cc, double con)
{
    complex \ r \,;
    r.re = cc.re * con;
    r.im = cc.im * con;
    return r;
// Other func
```

```
********************
// Append [seq_1] & [seq_2] to [seq_1, seq_2]
complex* append\_seq(complex seq\_1[], complex seq\_2[], int N) {
   complex* total_seq = new complex[N*2];
   for (int i = 0; i < N; i++) {
       total\_seq[i] = seq\_1[i];
   for (int i = N; i < 2*N; i++) {
       total\_seq[i] = seq\_2[i-N];
   return total_seq;
}
// Reorder the input_seq to an order
complex* reorder_seq(complex input_seq[], int N) {
   cout << "Reorder the sequence ..." << endl;
   complex* reordered_seq = new complex[N];
   for (int i = 0; i < N; ++i)
       int k = reverse\_bit(i, log2(N));
       reordered_seq[k] = input_seq[i];
   return reordered_seq;
}
      ***********************************
// Reverse Bit
   // input:
      // a decimal num,
       // N-based reverse method
   // output: a decimal num
int reverse_bit(int value, int N) {
   int ret = 0;
   int i = 0;
   while (i < N) {
       ret \ll 1;
       ret |= (value>>i) & 1;
       i++;
   return ret;
}
```

3.5 dif_fft.h — FFT implementation

```
// Work by Lin , Tzu-Heng
// W42, 2014011054
// Dept. of Electronic Engineering, Tsinghua University
// DSP Course Work
complex*\ DIF\_FFT\_reordered(complex\ input\_seq[]\ ,\ int\ N)\ ;
complex* DIF_FFT(complex input_seq[], int N, complex WN[], int recur_time_count);
         ************************
// DIF-FFT
   // input_seq[]:
   // N: size of input_seq
       // Must be a 2<sup>k</sup> integer
complex* DIF_FFT_reordered(complex input_seq[], int N) {
   // Initialize
   complex* reordered_seq = new complex[N];
   // Calc WN
   complex* WN = new complex[N];
   WN = Calc_WN(N);
   // Calc DIF-FFT
    reordered_seq = DIF_FFT(input_seq, N, WN, 0);
    // Reorder
    reordered_seq = reorder_seq(reordered_seq, N);
    return reordered_seq;
complex* DIF_FFT(complex input_seq[], int N, complex WN[], int recur_time_count) {
```

```
// cout << "\tDIF_FFT executed!\n"; // for validation
    // output seq
    complex* return_seq = new complex[N];
    if ( N != 2 ) {
        complex* first_half_seq = new complex[N/2];
        complex* second\_half\_seq = new complex[N/2];
        int k = pow(2, recur_time_count);
        // Calc
        for (int i = 0; i < N/2; ++i) {
            first_half_seq[i] = ComplexAdd(input_seq[i], input_seq[i+N/2]);
        for (int i = 0; i < N/2; ++i) {
            second_half_seq[i] = ComplexMul( ComplexAdd(input_seq[i], ReverseComplex
                (input\_seq[i+N/2])), WN[i*k]);
        }
        // DFT
        complex* DFTed_first_half_seq = new complex [N/2];
        DFTed_first_half_seq = DIF_FFT(first_half_seq, N/2, WN, recur_time_count+1);
        complex* DFTed\_second\_half\_seq = new complex [N/2];
        DFTed_second_half_seq = DIF_FFT(second_half_seq, N/2, WN, recur_time_count
            +1);
        // Append [DFTed_first_half_seq] & [DFTed_second_half_seq]
        return_seq = append_seq(DFTed_first_half_seq, DFTed_second_half_seq, N/2);
        return return_seq;
    \} else if ( N = 2 ) \{ // Smallest Butterfly Unit
        // cout << "\tDIF_FFT N==2 triggered!\n"; // for validation
        return_seq[0] = ComplexAdd(input_seq[0], input_seq[1]);
        return_seq[1] = ComplexMul( ComplexAdd(input_seq[0], ReverseComplex(
            input\_seq[1])), WN[0]);
        return return_seq;
   }
    // return [return_seq] # unordered
    return return_seq;
}
```

3.6 goertzel.h — Goertzel implementation

```
// Work by Lin, Tzu-Heng
```

```
// W42, 2014011054
// Dept. of Electronic Engineering, Tsinghua University
// DSP Course Work
// input x[n], N
// output: amp of the targeted 8 freq (freqs)
double* Goertzel(complex input_seq[], int N) {
    cout << "Calculating Goertzel ..." << endl;
    // targeted freqs
    double freqs[] =
                                              {697,
                                              770,
                                              852,
                                              941,
                         1209,1336,1477,1633};
    // Calc WN
    complex* WN = new complex[N];
    WN = Calc_WN(N);
    int sampling_rate = 8000;
    // Calc DFT of targeted 8 freqs
    complex targeted_X[8];
    complex* v = new complex[N];
    for (int i = 0; i < 8; ++i)
    {
        int k = ( N * freqs[i] ) / sampling_rate;
        double w_k = 2 * PI * k / N;
        // init
        v[0] = input\_seq[0];
        v[1] = ComplexAdd(input\_seq[1], ComplexScalarMul(v[0], 2*cos(w_k)));
        for (int j = 2; j < N; ++j)
            v[j] = ComplexAdd(ComplexAdd(input\_seq[j], ComplexScalarMul(v[j-1], 2*))
                cos(w_k)), ReverseComplex(v[j-2]);
        }
        targeted_X[i] = ComplexAdd(v[N-1], ReverseComplex(ComplexMul(WN[k], v[N-2]))
            );
    }
    // Calc amp
    double* amp_targeted_X = new double [8];
    for (int i = 0; i < 8; ++i)
        amp_targeted_X[i] = pow(targeted_X[i].re,2) + pow(targeted_X[i].im,2);
        // cout << amp_targeted_X[i] << endl; // indicator
```

```
return amp_targeted_X;
}
```

$3.7 \quad find_dtmf_symbol.h -- judge \ signals$

```
// Work by Lin, Tzu-Heng
// W42, 2014011054
// Dept. of Electronic Engineering, Tsinghua University
// DSP Course Work
// Decipher
char find_dtmf_symbol(double a, double b)
                    char symbol []
                          '4', '5', '6', 'B',
                          '7', '8', '9', 'C',
                          '*', '0', '#', 'D'};
    double freqs1[] =
                                              \{697,
                                               770,
                                              852,
                                               941};
    double freqs2[] = {1209,1336,1477,1633};
    // switch if a > b, making sure a < b
    if (a > b) {
        int c = a;
        a = b;
        b = c;
    }
    // get a pos: x
    int x = 100;
    for (int i = 0; i < 4; ++i)
        if~(~abs(freqs1\left[\,i\,\right]~-~a) <=~10~)
        {
            x = i;
            break;
        }
    }
    // get b pos: y
    int y = 100;
    for (int i = 0; i < 4; ++i)
        if ( abs(freqs2[i] - b) <= 10 )
        {
            y = i;
```

```
break;
}

if (x == 100 || y == 100)
{
    char error_char = 'x';
    return error_char;
}

// get return symbol
    char return_symbol = symbol[ x*4 + y ];

return return_symbol;
}
```

3.8 read wav.m — Convert .wav files to .txt

```
// Work by Lin, Tzu-Heng
// W42, 2014011054
// Dept. of Electronic Engineering, Tsinghua University
// DSP Course Work
close all; clc; clear;
%% Data1
dirpath = './Data1/';
writepath = './txtData1/';
dat = dir( dirpath );
for j = 1: length ( dat )
    if( ~isequal( dat( j ).name, '.') &...
        \simisequal( dat( j ).name, '..'))
        datpath = [dirpath dat( j ).name];
        [y, Fs] = audioread(datpath);
        f = fopen([writepath dat(j).name(1:8)'.txt'], 'w');
        fprintf(f, '\%f \ ', y);
        fclose(f);
    end
end
% Data2
datpath = './Data2/data.wav';
[y, Fs] = audioread(datpath);
f = fopen('./txtData2/data.txt', 'w');
fprintf(f, '\%f \ ', y);
fclose(f);
```