

7. SUMMARY AND CONCLUSION

The stethoscope remains the centerpiece of diagnostic tools for clinicians performing cardiac examinations of their patients. Many cardiac and pulmonary abnormalities can be diagnosed at the bedside with the use of the stethoscope. Auscultation is one of the most cost-effective diagnostic tests for patients with cardiac and pulmonary abnormalities.

Digital stethoscopes are designed to overcome the disadvantages of the acoustic stethoscopes. Digital stethoscope electronically amplifies body sounds. Digital stethoscope uses sound waves, which are converted from analog to electrical signals that are amplified and provide much louder and clearer sound. This makes identifying problems quicker and easier, and it allows the recording and playback of sounds picked up, making it useful for references or in the teaching for students.

In the present work, digital stethoscope system was designed and constructed. It consists of a function generator circuit for generating a sound signal with appropriate frequency and waveform to the constructed lung model by a mini-speaker inserted into the simulated trachea. Electrets condenser microphone at the surface of the model receives the sound signal and converts it into an electric signal. The signal then was amplified and filtered with a suitable gain and transmitted through an electrical cable to the sound card of a personal computer for further monitoring and analysis of the output sound signal.

The results of the simulated model showed that:

- The amplitude of the output sound signal increases with the increasing of solutions (Water, Sodium chloride, Glucose), and gel quantity.
- The sound attenuation decreases with the increasing of solutions quantity.
- The coefficient of attenuation decreases with the increasing of solutions quantity.

Digital stethoscope array system was developed for monitoring and diagnosing obstructive lung diseases which are associated with accumulation of lung fluids and cardiovascular diseases which associated with heart murmur like systolic and diastolic murmur from different auscultation points. Parameters such as sound attenuation and attenuation coefficient

The results of heart sound analysis showed that:

- The frequency spectrums of all normal and abnormal heart sounds were 64 Hz.
- The amplitude of normal heart sound at the apex auscultation point was -17 dB, while at the aortic and pulmonic auscultation points were -10 dB.
- The amplitude of abnormal heart sounds S3 and S4 at the apex auscultation point were -18dB.
- The amplitude of split S1 at the apex auscultation point was -22 dB, while the amplitude of split S2 at the pulmonic auscultation point was -16 dB.

- The amplitude of early, mid, late and holo systolic murmur at the apex auscultation point were in the range of -24 dB to -22 dB.

The results of lung sound analysis showed that:

- The amplitude of all normal and abnormal lung sounds was in the range of -50 dB to -30 dB.
- The frequency spectrum of normal vesicular lung sound was 120 Hz.
- The frequency spectrum of coarse crackles sound was 330 Hz.
- The frequency spectrum of inspiratory stridor sound was 275 Hz.
- The frequency spectrum of pleural friction sound was 58 Hz.
- The frequency spectrum of wheezing sound was 198 Hz.

8. REFERENCES

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FFT

FFT.cpp

```
*****//*
```

```
\file FFT.cpp
```

```
\brief Fast Fourier Transform routines.
```

This file contains a few FFT routines, including a real-FFT routine that is almost twice as fast as a normal complex FFT, and a power spectrum routine when you know you don't care about phase information.

Some of this code was based on a free implementation of an FFT by Don Cross, available on the web at:

<http://www.intersrv.com/~dcross/fft.html>

The basic algorithm for his code was based on Numerical Recipes in Fortran. I optimized his code further by reducing array accesses, caching the bit reversal table, and eliminating float-to-double conversions, and I added the routines to calculate a real FFT and a real power spectrum.

```
*/*****/  
/*
```

```
Added more window functions:
```

- * 4: Blackman
- * 5: Blackman-Harris
- * 6: Welch
- * 7: Gaussian(a=2.5)
- * 8: Gaussian(a=3.5)
- * 9: Gaussian(a=4.5)

```
*/
```

```
#include <wx/intl.h>  
#include <stdlib.h>  
#include <stdio.h>  
#include <math.h>
```

```
#include "FFT.h"
```

```
static int **gFFTBitTable = NULL;  
static const int MaxFastBits = 16;
```

```

/* Declare Static functions */
static int IsPowerOfTwo(int x);
static int NumberOfBitsNeeded(int PowerOfTwo);
static int ReverseBits(int index, int NumBits);
static void InitFFT();

int IsPowerOfTwo(int x)
{
    if (x < 2)
        return false;

    if (x & (x - 1))
        return false;

    return true;
}

int NumberOfBitsNeeded(int PowerOfTwo)
{
    int i;

    if (PowerOfTwo < 2) {
        fprintf(stderr, "Error: FFT called with size %d\n", PowerOfTwo);
        exit(1);
    }

    for (i = 0; i++;
        if (PowerOfTwo & (1 << i))
            return i;
    }

int ReverseBits(int index, int NumBits)
{
    int i, rev;

    for (i = rev = 0; i < NumBits; i++) {
        rev = (rev << 1) | (index & 1);
        index >>= 1;
    }

    return rev;
}

void InitFFT()
{
    gFFTBitTable = new int *[MaxFastBits];

    int len = 2;

```

```

for (int b = 1; b <= MaxFastBits; b++) {

    gFFTBitTable[b - 1] = new int[len];

    for (int i = 0; i < len; i++)
        gFFTBitTable[b - 1][i] = ReverseBits(i, b);

    len <<= 1;
}

#ifdef EXPERIMENTAL_USE_REALFFTf
#include "RealFFTf.h"
#endif

void DeinitFFT()
{
    if (gFFTBitTable) {
        for (int b = 1; b <= MaxFastBits; b++) {
            delete[] gFFTBitTable[b-1];
        }
        delete[] gFFTBitTable;
    }
#ifdef EXPERIMENTAL_USE_REALFFTf
    // Deallocate any unused RealFFTf tables
    CleanupFFT();
#endif
}

inline int FastReverseBits(int i, int NumBits)
{
    if (NumBits <= MaxFastBits)
        return gFFTBitTable[NumBits - 1][i];
    else
        return ReverseBits(i, NumBits);
}

/*
 * Complex Fast Fourier Transform
 */

void FFT(int NumSamples,
         bool InverseTransform,
         float *Realln, float *ImagIn, float *RealOut, float *ImagOut)
{
    int NumBits;          /* Number of bits needed to store indices */
    int i, j, k, n;
    int BlockSize, BlockEnd;

```

```

double angle_numerator = 2.0 * M_PI;
double tr, ti;          /* temp real, temp imaginary */

if (!IsPowerOfTwo(NumSamples)) {
    fprintf(stderr, "%d is not a power of two\n", NumSamples);
    exit(1);
}

if (!lgFFTBitTable)
    InitFFT();

if (!InverseTransform)
    angle_numerator = -angle_numerator;

NumBits = NumberOfBitsNeeded(NumSamples);

/*
** Do simultaneous data copy and bit-reversal ordering into outputs...
*/

for (i = 0; i < NumSamples; i++) {
    j = FastReverseBits(i, NumBits);
    RealOut[j] = RealIn[i];
    ImagOut[j] = (ImagIn == NULL) ? 0.0 : ImagIn[i];
}

/*
** Do the FFT itself...
*/

BlockEnd = 1;
for (BlockSize = 2; BlockSize <= NumSamples; BlockSize <<= 1) {

    double delta_angle = angle_numerator / (double) BlockSize;

    double sm2 = sin(-2 * delta_angle);
    double sm1 = sin(-delta_angle);
    double cm2 = cos(-2 * delta_angle);
    double cm1 = cos(-delta_angle);
    double w = 2 * cm1;
    double ar0, ar1, ar2, ai0, ai1, ai2;

    for (i = 0; i < NumSamples; i += BlockSize) {
        ar2 = cm2;
        ar1 = cm1;

        ai2 = sm2;

```

```

ai1 = sm1;

for (j = i, n = 0; n < BlockEnd; j++, n++) {
    ar0 = w * ar1 - ar2;
    ar2 = ar1;
    ar1 = ar0;

    ai0 = w * ai1 - ai2;
    ai2 = ai1;
    ai1 = ai0;

    k = j + BlockEnd;
    tr = ar0 * RealOut[k] - ai0 * ImagOut[k];
    ti = ar0 * ImagOut[k] + ai0 * RealOut[k];

    RealOut[k] = RealOut[j] - tr;
    ImagOut[k] = ImagOut[j] - ti;

    RealOut[j] += tr;
    ImagOut[j] += ti;
}
}

BlockEnd = BlockSize;
}

/*
** Need to normalize if inverse transform...
*/

if (InverseTransform) {
    float denom = (float) NumSamples;

    for (i = 0; i < NumSamples; i++) {
        RealOut[i] /= denom;
        ImagOut[i] /= denom;
    }
}

/*
* Real Fast Fourier Transform
*
* This function was based on the code in Numerical Recipes in C.
* In Num. Rec., the inner loop is based on a single 1-based array
* of interleaved real and imaginary numbers. Because we have two
* separate zero-based arrays, our indices are quite different.
* Here is the correspondence between Num. Rec. indices and our indices:

```

```

*
* i1 <-> real[i]
* i2 <-> imag[i]
* i3 <-> real[n/2-i]
* i4 <-> imag[n/2-i]
*/

void RealFFT(int NumSamples, float *RealIn, float *RealOut, float *ImagOut)
{
#ifdef EXPERIMENTAL_USE_REALFFT
    // Remap to RealFFTf() function
    int i;
    HFFT hFFT = GetFFT(NumSamples);
    float *pFFT = new float[NumSamples];
    // Copy the data into the processing buffer
    for(i=0; i<NumSamples; i++)
        pFFT[i] = RealIn[i];

    // Perform the FFT
    RealFFTf(pFFT, hFFT);

    // Copy the data into the real and imaginary outputs
    for(i=1; i<(NumSamples/2); i++) {
        RealOut[i]=pFFT[hFFT->BitReversed[i] ];
        ImagOut[i]=pFFT[hFFT->BitReversed[i]+1];
    }
    // Handle the (real-only) DC and Fs/2 bins
    RealOut[0] = pFFT[0];
    RealOut[1] = pFFT[1];
    ImagOut[0] = ImagOut[1] = 0;
    // Fill in the upper half using symmetry properties
    for(i++; i<NumSamples; i++) {
        RealOut[i] = RealOut[NumSamples-i];
        ImagOut[i] = -ImagOut[NumSamples-i];
    }
    delete [] pFFT;
    ReleaseFFT(hFFT);
#else

    int Half = NumSamples / 2;
    int i;

    float theta = M_PI / Half;

    float *tmpReal = new float[Half];
    float *tmpImag = new float[Half];

```

```

for (i = 0; i < Half; i++) {
    tmpReal[i] = Realln[2 * i];
    tmpImag[i] = Realln[2 * i + 1];
}

FFT(Half, 0, tmpReal, tmpImag, RealOut, ImagOut);

float wtemp = float (sin(0.5 * theta));

float wpr = -2.0 * wtemp * wtemp;
float wpi = -1.0 * float (sin(theta));
float wr = 1.0 + wpr;
float wi = wpi;

int i3;

float h1r, h1i, h2r, h2i;

for (i = 1; i < Half / 2; i++) {

    i3 = Half - i;

    h1r = 0.5 * (RealOut[i] + RealOut[i3]);
    h1i = 0.5 * (ImagOut[i] - ImagOut[i3]);
    h2r = 0.5 * (ImagOut[i] + ImagOut[i3]);
    h2i = -0.5 * (RealOut[i] - RealOut[i3]);

    RealOut[i] = h1r + wr * h2r - wi * h2i;
    ImagOut[i] = h1i + wr * h2i + wi * h2r;
    RealOut[i3] = h1r - wr * h2r + wi * h2i;
    ImagOut[i3] = -h1i + wr * h2i + wi * h2r;

    wr = (wtemp = wr) * wpr - wi * wpi + wr;
    wi = wi * wpr + wtemp * wpi + wi;
}

RealOut[0] = (h1r = RealOut[0]) + ImagOut[0];
ImagOut[0] = h1r - ImagOut[0];

delete[]tmpReal;
delete[]tmpImag;
#endif //EXPERIMENTAL_USE_REALFFTF
}

#ifdef EXPERIMENTAL_USE_REALFFTF
/*
* InverseRealFFT
*

```

```

* This function computes the inverse of RealFFT, above.
* The RealIn and ImagIn is assumed to be conjugate-symmetric
* and as a result the output is purely real.
* Only the first half of RealIn and ImagIn are used due to this
* symmetry assumption.
*/
void InverseRealFFT(int NumSamples, float *RealIn, float *ImagIn, float *RealOut)
{
    // Remap to RealFFTf() function
    int i;
    HFFT hFFT = GetFFT(NumSamples);
    float *pFFT = new float[NumSamples];
    // Copy the data into the processing buffer
    for(i=0; i<(NumSamples/2); i++)
        pFFT[2*i ] = RealIn[i];
    if(ImagIn == NULL) {
        for(i=0; i<(NumSamples/2); i++)
            pFFT[2*i+1] = 0;
    } else {
        for(i=0; i<(NumSamples/2); i++)
            pFFT[2*i+1] = ImagIn[i];
    }
    // Put the fs/2 component in the imaginary part of the DC bin
    pFFT[1] = RealIn[i];

    // Perform the FFT
    InverseRealFFTf(pFFT, hFFT);

    // Copy the data to the (purely real) output buffer
    ReorderToTime(hFFT, pFFT, RealOut);

    delete [] pFFT;
    ReleaseFFT(hFFT);
}
#endif // EXPERIMENTAL_USE_REALFFT

/*
* PowerSpectrum
*
* This function computes the same as RealFFT, above, but
* adds the squares of the real and imaginary part of each
* coefficient, extracting the power and throwing away the
* phase.
*
* For speed, it does not call RealFFT, but duplicates some
* of its code.
*/

```

```

void PowerSpectrum(int NumSamples, float *In, float *Out)
{
#ifdef EXPERIMENTAL_USE_REALFFTF
    // Remap to RealFFTF() function
    int i;
    HFFT hFFT = GetFFT(NumSamples);
    float *pFFT = new float[NumSamples];
    // Copy the data into the processing buffer
    for(i=0; i<NumSamples; i++)
        pFFT[i] = In[i];

    // Perform the FFT
    RealFFTF(pFFT, hFFT);

    // Copy the data into the real and imaginary outputs
    for(i=1; i<NumSamples/2; i++) {
        Out[i] = (pFFT[hFFT->BitReversed[i]] * pFFT[hFFT->BitReversed[i]])
            + (pFFT[hFFT->BitReversed[i]+1] * pFFT[hFFT->BitReversed[i]+1]);
    }
    // Handle the (real-only) DC and Fs/2 bins
    Out[0] = pFFT[0] * pFFT[0];
    Out[i] = pFFT[1] * pFFT[1];
    delete [] pFFT;
    ReleaseFFT(hFFT);
#else // EXPERIMENTAL_USE_REALFFTF

    int Half = NumSamples / 2;
    int i;

    float theta = M_PI / Half;

    float *tmpReal = new float[Half];
    float *tmpImag = new float[Half];
    float *RealOut = new float[Half];
    float *ImagOut = new float[Half];

    for (i = 0; i < Half; i++) {
        tmpReal[i] = In[2 * i];
        tmpImag[i] = In[2 * i + 1];
    }

    FFT(Half, 0, tmpReal, tmpImag, RealOut, ImagOut);

    float wtemp = float (sin(0.5 * theta));

    float wpr = -2.0 * wtemp * wtemp;
    float wpi = -1.0 * float (sin(theta));

```

```

float wr = 1.0 + wpr;
float wi = wpi;

int i3;

float h1r, h1i, h2r, h2i, rt, it;

for (i = 1; i < Half / 2; i++) {

    i3 = Half - i;

    h1r = 0.5 * (RealOut[i] + RealOut[i3]);
    h1i = 0.5 * (ImagOut[i] - ImagOut[i3]);
    h2r = 0.5 * (ImagOut[i] + ImagOut[i3]);
    h2i = -0.5 * (RealOut[i] - RealOut[i3]);

    rt = h1r + wr * h2r - wi * h2i;
    it = h1i + wr * h2i + wi * h2r;

    Out[i] = rt * rt + it * it;

    rt = h1r - wr * h2r + wi * h2i;
    it = -h1i + wr * h2i + wi * h2r;

    Out[i3] = rt * rt + it * it;

    wr = (wtemp = wr) * wpr - wi * wpi + wr;
    wi = wi * wpr + wtemp * wpi + wi;
}

rt = (h1r = RealOut[0]) + ImagOut[0];
it = h1r - ImagOut[0];
Out[0] = rt * rt + it * it;

rt = RealOut[Half / 2];
it = ImagOut[Half / 2];
Out[Half / 2] = rt * rt + it * it;

delete[]tmpReal;
delete[]tmpImag;
delete[]RealOut;
delete[]ImagOut;
#endif // EXPERIMENTAL_USE_REALFFTF
}

/*
 * Windowing Functions
 */

```

```

int NumWindowFuncs()
{
    return 10;
}

const wxChar *WindowFuncName(int whichFunction)
{
    switch (whichFunction) {
    default:
    case 0:
        return _("Rectangular");
    case 1:
        return wxT("Bartlett");
    case 2:
        return wxT("Hamming");
    case 3:
        return wxT("Hanning");
    case 4:
        return wxT("Blackman");
    case 5:
        return wxT("Blackman-Harris");
    case 6:
        return wxT("Welch");
    case 7:
        return wxT("Gaussian(a=2.5)");
    case 8:
        return wxT("Gaussian(a=3.5)");
    case 9:
        return wxT("Gaussian(a=4.5)");
    }
}

void WindowFunc(int whichFunction, int NumSamples, float *in)
{
    int i;
    double A;

    switch( whichFunction )
    {
    case 1:
        // Bartlett (triangular) window
        for (i = 0; i < NumSamples / 2; i++) {
            in[i] *= (i / (float) (NumSamples / 2));
            in[i + (NumSamples / 2)] *=
                (1.0 - (i / (float) (NumSamples / 2)));
        }
        break;

```

```

case 2:
    // Hamming
    for (i = 0; i < NumSamples; i++)
        in[i] *= 0.54 - 0.46 * cos(2 * M_PI * i / (NumSamples - 1));
    break;
case 3:
    // Hanning
    for (i = 0; i < NumSamples; i++)
        in[i] *= 0.50 - 0.50 * cos(2 * M_PI * i / (NumSamples - 1));
    break;
case 4:
    // Blackman
    for (i = 0; i < NumSamples; i++) {
        in[i] *= 0.42 - 0.5 * cos (2 * M_PI * i / (NumSamples - 1)) + 0.08 * cos (4 * M_PI * i / (NumSamples -
1));
    }
    break;
case 5:
    // Blackman-Harris
    for (i = 0; i < NumSamples; i++) {
        in[i] *= 0.35875 - 0.48829 * cos(2 * M_PI * i / (NumSamples-1)) + 0.14128 * cos(4 * M_PI *
i / (NumSamples-1)) - 0.01168 * cos(6 * M_PI * i / (NumSamples-1));
    }
    break;
case 6:
    // Welch
    for (i = 0; i < NumSamples; i++) {
        in[i] *= 4*i/(float)NumSamples*(1-(i/(float)NumSamples));
    }
    break;
case 7:
    // Gaussian (a=2.5)
    // Precalculate some values, and simplify the fmla to try and reduce overhead
    A=-2*2.5*2.5;

    for (i = 0; i < NumSamples; i++) {
        // full
        // in[i] *= exp(-0.5*(A*((i-NumSamples/2)/NumSamples/2))*A*((i-
NumSamples/2)/NumSamples/2));
        // reduced
        in[i] *= exp(A*(0.25 + ((i/(float)NumSamples)*(i/(float)NumSamples)) - (i/(float)NumSamples)));
    }
    break;
case 8:
    // Gaussian (a=3.5)
    A=-2*3.5*3.5;
    for (i = 0; i < NumSamples; i++) {
        // reduced

```

```
        in[i] *= exp(A*(0.25 + ((i/(float)NumSamples)*(i/(float)NumSamples)) - (i/(float)NumSamples)));
    }
    break;
case 9:
    // Gaussian (a=4.5)
    A=-2*4.5*4.5;
    for (i = 0; i < NumSamples; i++) {
        // reduced
        in[i] *= exp(A*(0.25 + ((i/(float)NumSamples)*(i/(float)NumSamples)) - (i/(float)NumSamples)));
    }
    break;
default:
    fprintf(stderr,"FFT::WindowFunc - Invalid window function: %d\n",whichFunction);
}
}
```

LM741

Operational Amplifier

General Description

The LM741 series are general purpose operational amplifiers which feature improved performance over industry standards like the LM709. They are direct, plug-in replacements for the 709C, LM201, MC1439 and 748 in most applications. The amplifiers offer many features which make their application nearly foolproof: overload protection on the input and

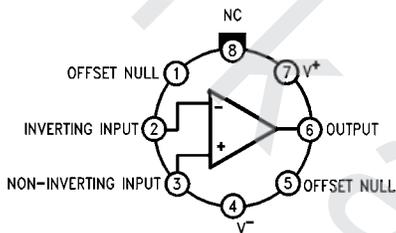
output, no latch-up when the common mode range is exceeded, as well as freedom from oscillations.

The LM741C is identical to the LM741/LM741A except that the LM741C has their performance guaranteed over a 0°C to +70°C temperature range, instead of -55°C to +125°C.

Features

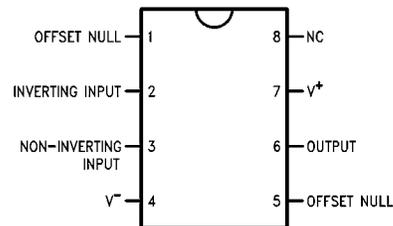
Connection Diagrams

Metal Can Package



00934102

Dual-In-Line or S.O. Package



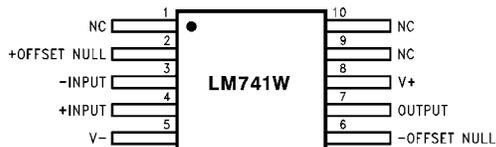
00934103

Note 1: LM741H is available per JM38510/10101

**Order Number LM741H, LM741H/883 (Note 1),
LM741AH/883 or LM741CH**
See NS Package Number H08C

Order Number LM741J, LM741J/883, LM741CN
See NS Package Number J08A, M08A or N08E

Ceramic Flatpak

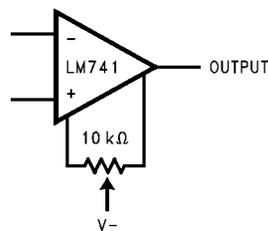


00934106

Order Number LM741W/883
See NS Package Number W10A

Typical Application

Offset Nulling Circuit



00934107

Absolute Maximum Ratings (Note 2)

If Military/Aerospace specified devices are required, please contact the National Semiconductor Sales Office/Distributors for availability and specifications.

(Note 7)

	LM741A	LM741	LM741C
Supply Voltage	±22V	±22V	±18V
Power Dissipation (Note 3)	500 mW	500 mW	500 mW
Differential Input Voltage	±30V	±30V	±30V
Input Voltage (Note 4)	±15V	±15V	±15V
Output Short Circuit Duration	Continuous	Continuous	Continuous
Operating Temperature Range	-55°C to +125°C	-55°C to +125°C	0°C to +70°C
Storage Temperature Range	-65°C to +150°C	-65°C to +150°C	-65°C to +150°C
Junction Temperature	150°C	150°C	100°C
Soldering Information			
N-Package (10 seconds)	260°C	260°C	260°C
J- or H-Package (10 seconds)	300°C	300°C	300°C
M-Package			
Vapor Phase (60 seconds)	215°C	215°C	215°C
Infrared (15 seconds)	215°C	215°C	215°C
See AN-450 "Surface Mounting Methods and Their Effect on Product Reliability" for other methods of soldering surface mount devices.			
ESD Tolerance (Note 8)	400V	400V	400V

Electrical Characteristics (Note 5)

Parameter	Conditions	LM741A			LM741			LM741C			Units
		Min	Typ	Max	Min	Typ	Max	Min	Typ	Max	
Input Offset Voltage	$T_A = 25^\circ\text{C}$ $R_S \leq 10\text{ k}\Omega$ $R_S \leq 50\Omega$		0.8	3.0		1.0	5.0		2.0	6.0	mV
	$T_{AMIN} \leq T_A \leq T_{AMAX}$ $R_S \leq 50\Omega$ $R_S \leq 10\text{ k}\Omega$			4.0			6.0			7.5	mV
Average Input Offset Voltage Drift				15							$\mu\text{V}/^\circ\text{C}$
Input Offset Voltage Adjustment Range	$T_A = 25^\circ\text{C}$, $V_S = \pm 20\text{V}$	±10				±15			±15		mV
Input Offset Current	$T_A = 25^\circ\text{C}$		3.0	30		20	200		20	200	nA
	$T_{AMIN} \leq T_A \leq T_{AMAX}$			70		85	500			300	nA
Average Input Offset Current Drift				0.5							$\text{nA}/^\circ\text{C}$
Input Bias Current	$T_A = 25^\circ\text{C}$		30	80		80	500		80	500	nA
	$T_{AMIN} \leq T_A \leq T_{AMAX}$			0.210			1.5			0.8	μA
Input Resistance	$T_A = 25^\circ\text{C}$, $V_S = \pm 20\text{V}$	1.0	6.0		0.3	2.0		0.3	2.0		$\text{M}\Omega$
	$T_{AMIN} \leq T_A \leq T_{AMAX}$, $V_S = \pm 20\text{V}$	0.5									$\text{M}\Omega$
Input Voltage Range	$T_A = 25^\circ\text{C}$							±12	±13		V
	$T_{AMIN} \leq T_A \leq T_{AMAX}$				±12	±13					V

Electrical Characteristics (Note 5) (Continued)

Parameter	Conditions	LM741A			LM741			LM741C			Units	
		Min	Typ	Max	Min	Typ	Max	Min	Typ	Max		
Large Signal Voltage Gain	$T_A = 25^\circ\text{C}$, $R_L \geq 2\text{ k}\Omega$ $V_S = \pm 20\text{V}$, $V_O = \pm 15\text{V}$ $V_S = \pm 15\text{V}$, $V_O = \pm 10\text{V}$	50			50	200		20	200		V/mV V/mV	
	$T_{AMIN} \leq T_A \leq T_{AMAX}$, $R_L \geq 2\text{ k}\Omega$, $V_S = \pm 20\text{V}$, $V_O = \pm 15\text{V}$ $V_S = \pm 15\text{V}$, $V_O = \pm 10\text{V}$	32			25			15			V/mV V/mV	
	$V_S = \pm 5\text{V}$, $V_O = \pm 2\text{V}$	10									V/mV	
Output Voltage Swing	$V_S = \pm 20\text{V}$ $R_L \geq 10\text{ k}\Omega$	± 16									V	
	$R_L \geq 2\text{ k}\Omega$	± 15									V	
	$V_S = \pm 15\text{V}$ $R_L \geq 10\text{ k}\Omega$				± 12	± 14		± 12	± 14		V	
	$R_L \geq 2\text{ k}\Omega$				± 10	± 13		± 10	± 13		V	
Output Short Circuit Current	$T_A = 25^\circ\text{C}$	10	25	35		25			25		mA	
	$T_{AMIN} \leq T_A \leq T_{AMAX}$	10		40							mA	
Common-Mode Rejection Ratio	$T_{AMIN} \leq T_A \leq T_{AMAX}$ $R_S \leq 10\text{ k}\Omega$, $V_{CM} = \pm 12\text{V}$				70	90		70	90		dB	
	$R_S \leq 50\Omega$, $V_{CM} = \pm 12\text{V}$	80	95								dB	
Supply Voltage Rejection Ratio	$T_{AMIN} \leq T_A \leq T_{AMAX}$, $V_S = \pm 20\text{V}$ to $V_S = \pm 5\text{V}$ $R_S \leq 50\Omega$	86	96								dB	
	$R_S \leq 10\text{ k}\Omega$				77	96		77	96		dB	
Transient Response	$T_A = 25^\circ\text{C}$, Unity Gain	Rise Time		0.25	0.8		0.3		0.3		μs	
		Overshoot		6.0	20		5		5		%	
Bandwidth (Note 6)	$T_A = 25^\circ\text{C}$	0.437	1.5								MHz	
Slew Rate	$T_A = 25^\circ\text{C}$, Unity Gain	0.3	0.7			0.5			0.5		V/ μs	
Supply Current	$T_A = 25^\circ\text{C}$					1.7	2.8		1.7	2.8	mA	
Power Consumption	$T_A = 25^\circ\text{C}$ $V_S = \pm 20\text{V}$ $V_S = \pm 15\text{V}$	LM741A		80	150						mW mW	
	$V_S = \pm 20\text{V}$ $T_A = T_{AMIN}$ $T_A = T_{AMAX}$	LM741			165							mW mW
					135							
$V_S = \pm 15\text{V}$ $T_A = T_{AMIN}$ $T_A = T_{AMAX}$	LM741					60	100				mW mW	
						45	75					

Note 2: "Absolute Maximum Ratings" indicate limits beyond which damage to the device may occur. Operating Ratings indicate conditions for which the device is functional, but do not guarantee specific performance limits.

Electrical Characteristics (Note 5) (Continued)

Note 3: For operation at elevated temperatures, these devices must be derated based on thermal resistance, and T_j max. (listed under "Absolute Maximum Ratings"). $T_j = T_A + (\theta_{JA} P_D)$.

Thermal Resistance	Cerdip (J)	DIP (N)	HO8 (H)	SO-8 (M)
θ_{JA} (Junction to Ambient)	100°C/W	100°C/W	170°C/W	195°C/W
θ_{JC} (Junction to Case)	N/A	N/A	25°C/W	N/A

Note 4: For supply voltages less than $\pm 15V$, the absolute maximum input voltage is equal to the supply voltage.

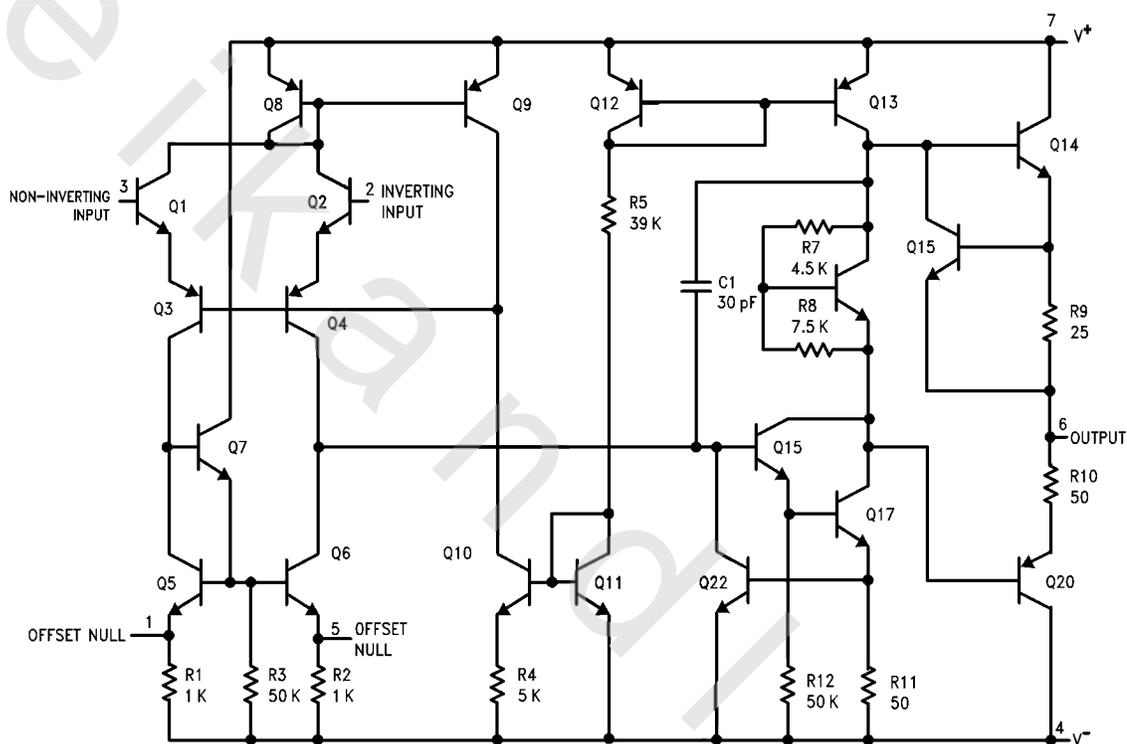
Note 5: Unless otherwise specified, these specifications apply for $V_S = \pm 15V$, $-55^\circ C \leq T_A \leq +125^\circ C$ (LM741/LM741A). For the LM741C/LM741E, these specifications are limited to $0^\circ C \leq T_A \leq +70^\circ C$.

Note 6: Calculated value from: BW (MHz) = $0.35/\text{Rise Time}(\mu s)$.

Note 7: For military specifications see RETS741X for LM741 and RETS741AX for LM741A.

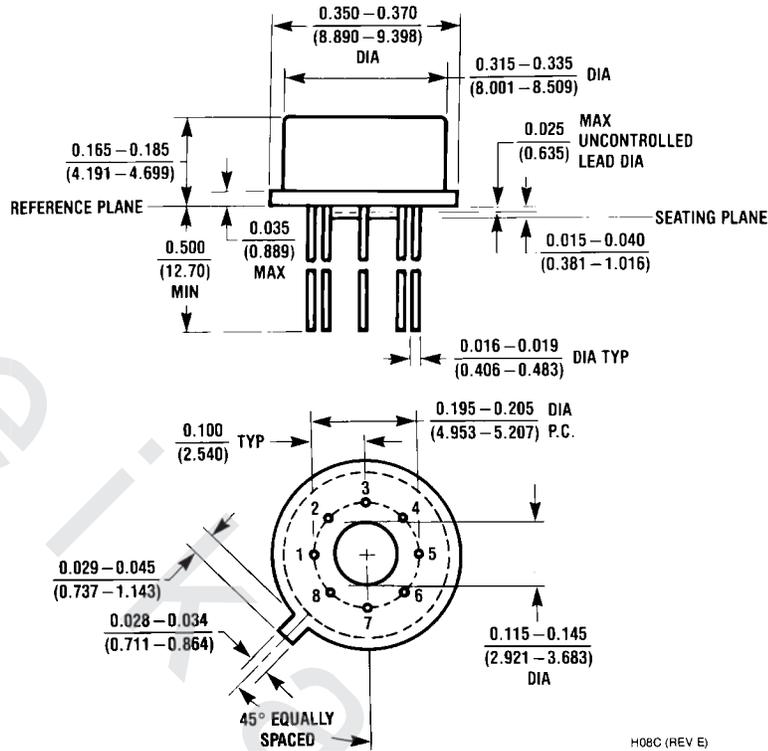
Note 8: Human body model, $1.5\text{ k}\Omega$ in series with 100 pF .

Schematic Diagram



Physical Dimensions inches (millimeters)

unless otherwise noted

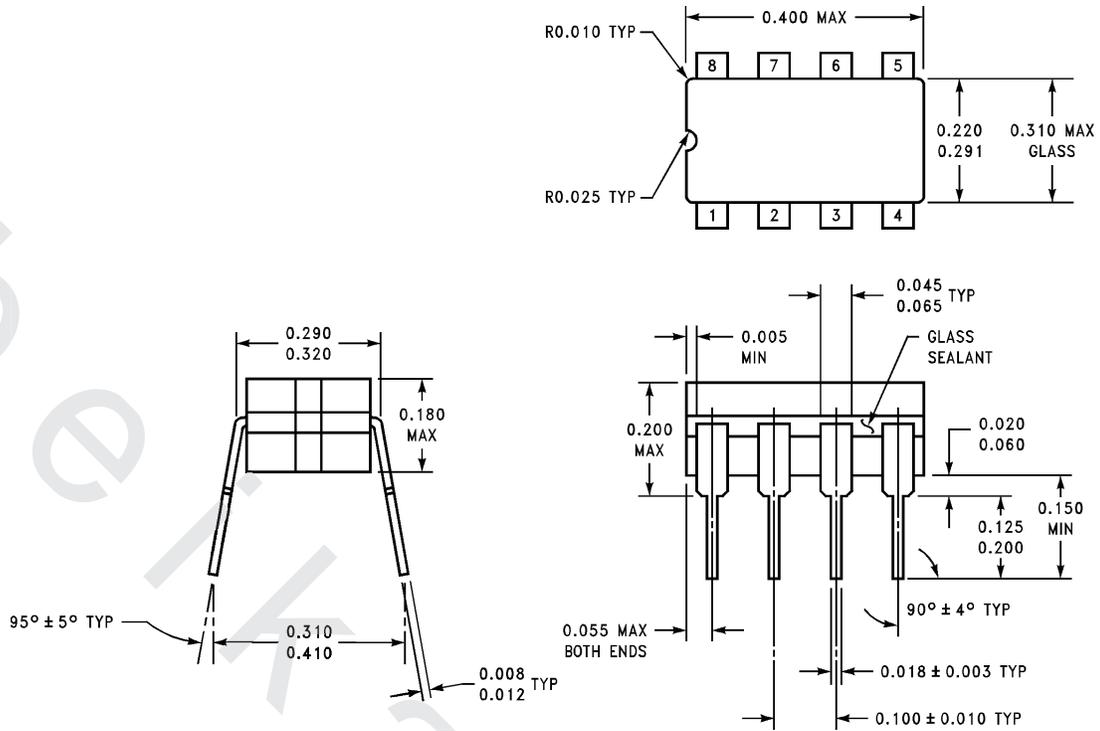


Metal Can Package (H)

Order Number LM741H, LM741H/883, LM741AH/883, LM741AH-MIL or LM741CH
NS Package Number H08C

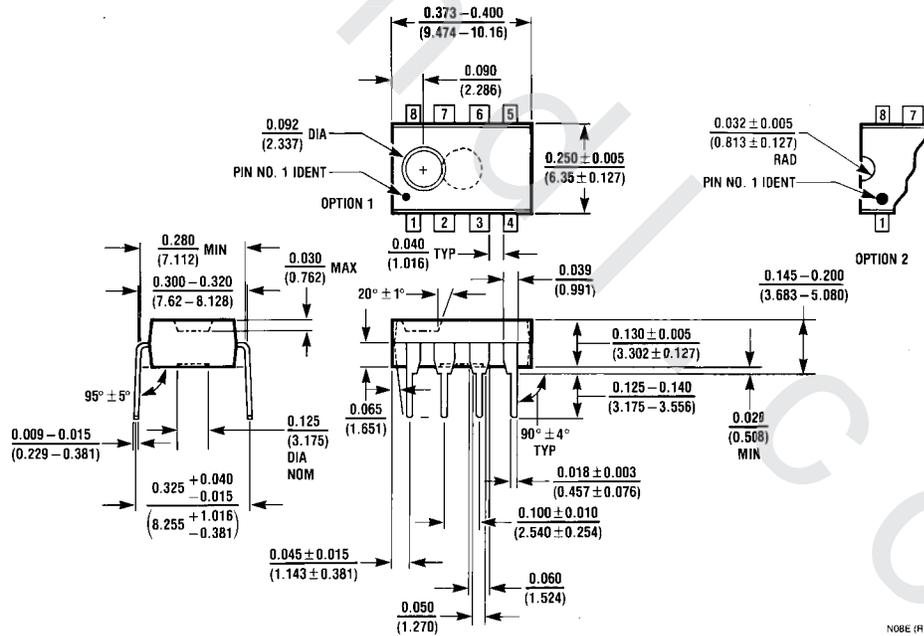
H08C (REV E)

Physical Dimensions inches (millimeters) unless otherwise noted (Continued)



J08A (REV K)

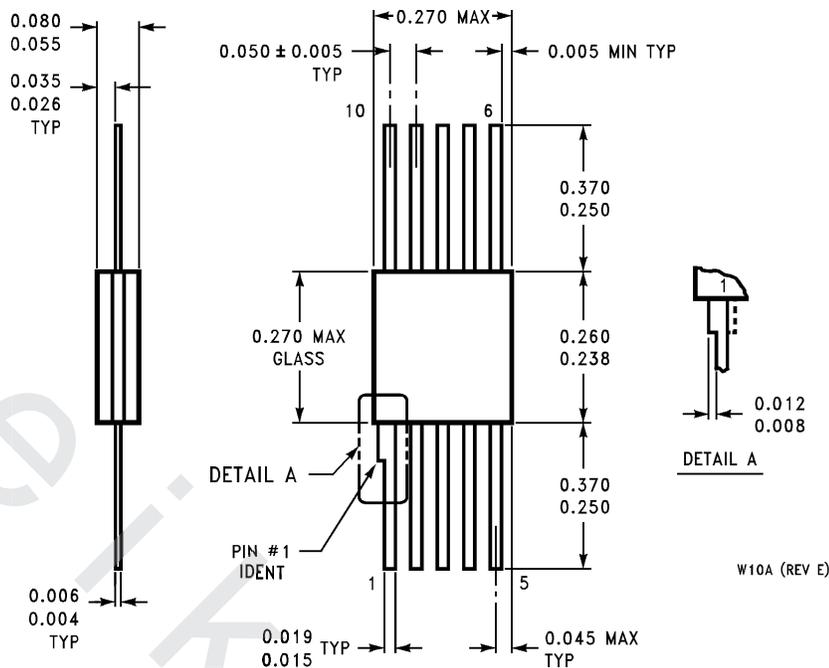
Ceramic Dual-In-Line Package (J)
Order Number LM741J/883
NS Package Number J08A



NO8E (REV F)

Dual-In-Line Package (N)
Order Number LM741CN
NS Package Number N08E

Physical Dimensions inches (millimeters) unless otherwise noted (Continued)



10-Lead Ceramic Flatpak (W)
Order Number LM741W/883, LM741WG-MPR or LM741WG/883
NS Package Number W10A

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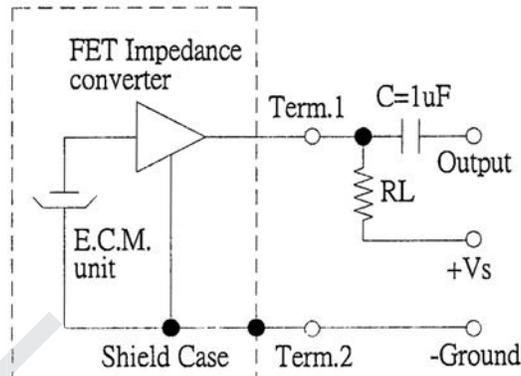
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PART NUMBER: CMA-4544PF-W

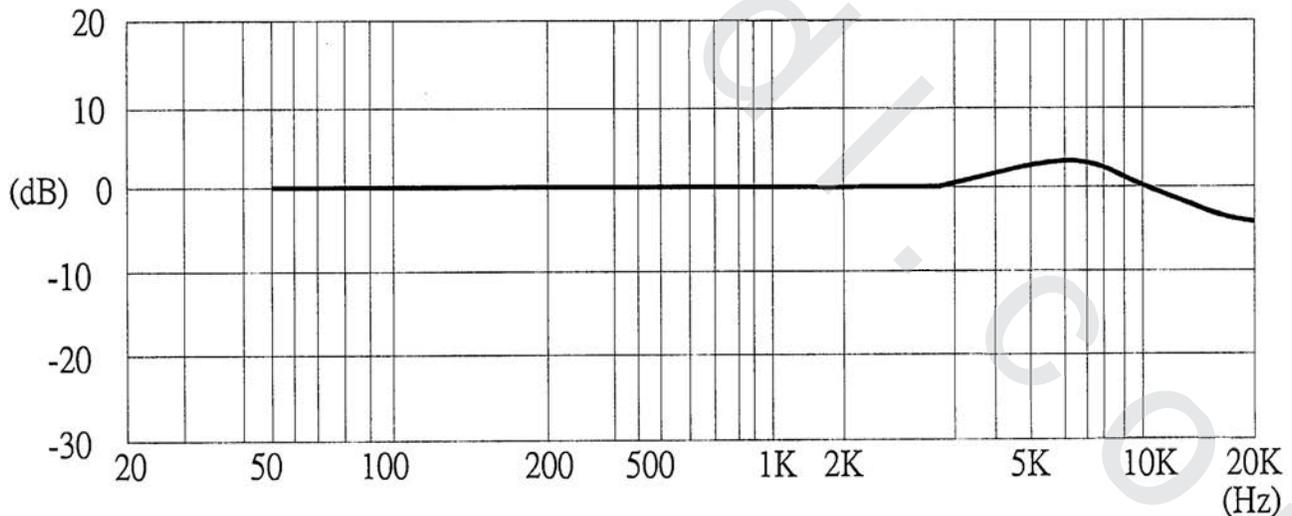
DESCRIPTION: electret condenser microphone

MEASUREMENT CIRCUIT



Schematic Diagram **RL=2.2KΩ**

FREQUENCY RESPONSE CURVE



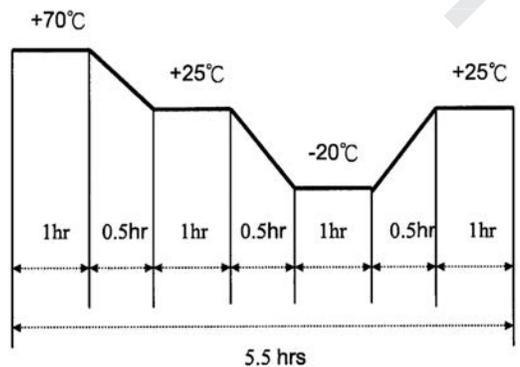
PART NUMBER: CMA-4544PF-W

DESCRIPTION: electret condenser microphone

MECHANICAL CHARACTERISTICS

item	test condition	evaluation standard
soldering heat resistance	Lead terminals are immersed in solder bath of $270 \pm 5^\circ\text{C}$ for 2 ± 0.5 seconds.	No interference in operation.
PCB wire pull strength	The pull force will be applied to double lead wire: Horizontal 4.9N (0.5kg) for 30 seconds	No damage or cutting off.
vibration	The part will be measured after applying a vibration amplitude of 1.5 mm with 10 to 55 Hz band of vibration frequency to each of the 3 perpendicular directions for 2 hours.	After any tests, the sensitivity should be within $\pm 3\text{dB}$ compared to the initial measurement.
drop test	The part will be dropped from a height of 1 m onto a 20 mm thick wooden board 3 times in 3 axes (X, Y, Z) for a total of 9 drops.	

ENVIRONMENT TEST

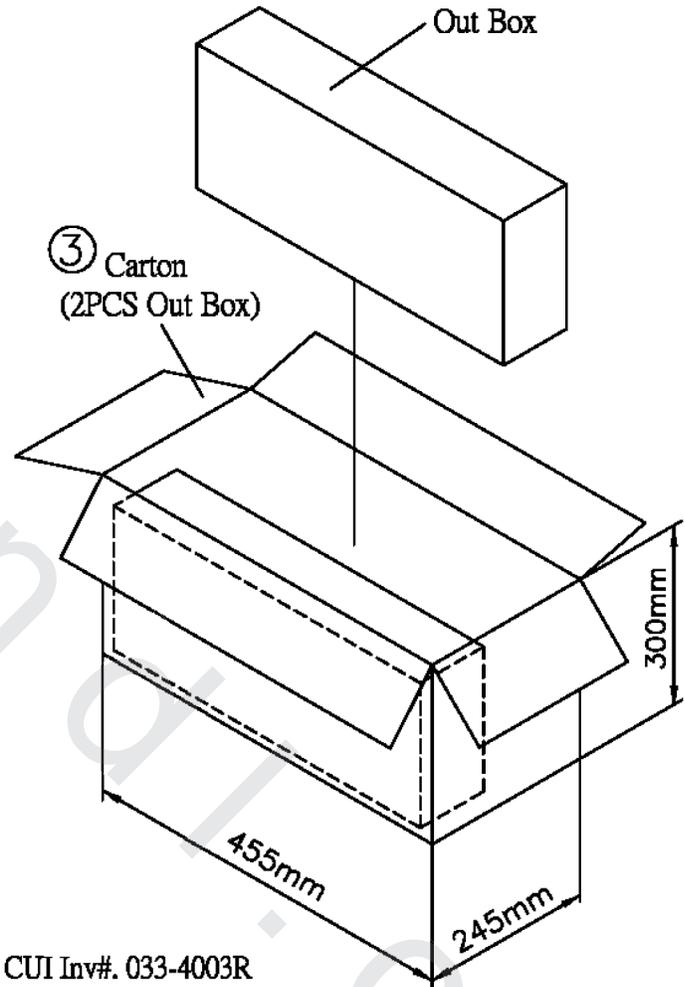
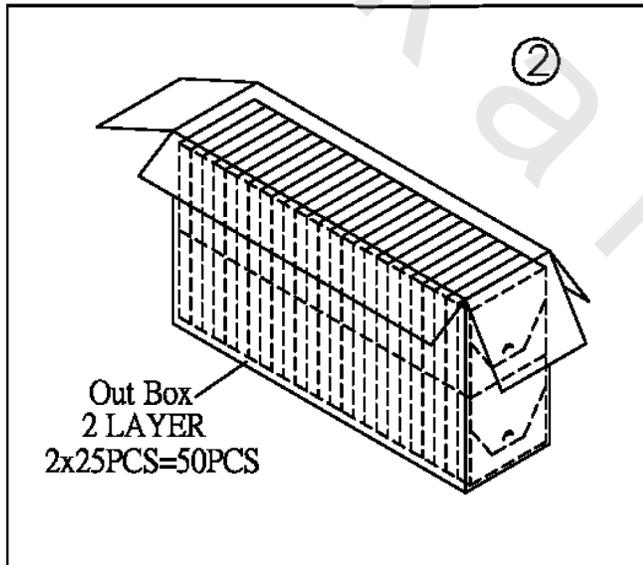
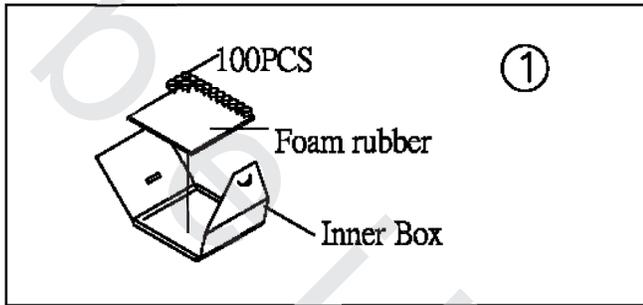
item	test condition	evaluation standard
high temp. test	After being placed in a chamber at $+70^\circ\text{C}$ for 72 hours.	The part will be measured after being placed at $+25^\circ\text{C}$ for 6 hours. After any tests, the sensitivity should be within $\pm 3\text{dB}$ compared to the initial measurement.
low temp. test	After being placed in a chamber at -20°C for 72 hours.	
humidity test	After being placed in a chamber at $+40^\circ\text{C}$ and $90 \pm 5\%$ relative humidity for 240 hours.	
temp. cycle test	The part shall be subjected to 10 cycles. One cycle will consist of: 	

TEST CONDITIONS

standard test condition	a) temperature: $+5 \sim +35^\circ\text{C}$	b) humidity: 45 - 85%	c) pressure: 860-1060 mbar
judgement test condition	a) temperature: $+25 \pm 2^\circ\text{C}$	b) humidity: 60 - 70%	c) pressure: 860-1060 mbar

PART NUMBER: CMA-4544PF-W

DESCRIPTION: electret condenser microphone

PACKAGING


1. CUI Inv#. 033-4003R
CUI Part#. CMA-4544PF-W
2. RoHS Compliant

Inner Box	100mmx100mmx15mm	100PCSx1=100PCS
Out Box	435mmx120mmx280mm	100PCSx50=5,000PCS
Carton Box	455mmx245mmx300mm	5,000PCSx2=10,000PCS

الملخص العربي

تبقى سماعة الطبيب محور أدوات التشخيص بالنسبة للأطباء لأداء امتحانات القلب لمرضاهم. يمكن تشخيص العديد من التشوهات القلبية والرئوية مع استخدام سماعة الطبيب. التسمع هو واحد من الاختبارات التشخيصية الأكثر فعالية من حيث التكلفة للمرضى الذين يعانون من تشوهات قلبية ورئوية.

صممت السماعات الرقمية للتغلب على عيوب السماعات الصوتية. سماعة الطبيب الرقمية تضخم الأصوات إلكترونياً. يستخدم سماعة الموجات الصوتية الرقمية، والتي يتم تحويلها من التناظرية إلى إشارات كهربائية يتم تضخيمها وتوفير الكثير من لودر وأوضح الصوت. وهذا يجعل تحديد المشاكل بشكل أسرع وأسهل، وأنه يسمح اختار تسجيل وتشغيل الأصوات تصل، مما يجعلها مفيدة للمراجع أو في التدريس للطلاب.

في العمل الحالي، تم تصميم نظام سماعة الطبيب الرقمية والتي شيدت للتغلب على القيود المفروضة على السماعات التقليدية. وهو يتألف من دائرة مولد إشارة لتوليد إشارة الصوت مع تردد والموجي المناسب للنموذج التي شيدت من قبل المتكلم مصغرة. مكثف ميكروفون electrets يتلقى إشارة الصوت وتحويلها إلى إشارة كهربائية. إشارة ثم تم تضخيمها وتصفيتها مع زيادة مناسبة والتي تنتقل عن طريق كابل كهربائي لبطاقة الصوت من جهاز كمبيوتر شخصي (PC) لمزيد من الرصد والتحليل للإشارة الصوت الناتج.

أظهرت نتائج النموذج محاكاة ما يلي:

- السعة من الصوت الناتج إشارة زيادات مع زيادة من الحول (المياه، وكلوريد الصوديوم، الجلوكوز)، وكمية الجل.
- وتوهين الصوت ومعامل التوهين يتناقص مع زيادة كمية من الحول.

وقد تم تطوير نظام رقمي مجموعة سماعة الطبيب لرصد وتشخيص أمراض الرئة الانسدادي التي ترتبط مع تراكم السوائل في الرئة والأمراض القلبية الوعائية التي ترتبط مع نفخة القلب مثل نفخة الانقباضي والانقباضي من النقاط التسمع مختلفة.

وأظهرت نتائج تحليل سليم القلب ما يلي:

- وكانت أطراف التردد لجميع الأصوات القلب العادية وغير العادية ٦٤ هرتز.
- كان اتساع صوت القلب الطبيعي عند نقطة التسمع قمة -١٧ ديسيبل، بينما في نقاط التسمع الأبهري والرئوي كانت -١٠ ديسيبل.
- واتساع القلب غير طبيعية الأصوات كانت S3 S4 وعند نقطة التسمع قمة
- كان اتساع الانقسام S1 عند نقطة التسمع قمة -٢٢ ديسيبل، بينما كان اتساع الانقسام S2 عند نقطة التسمع الرئوي -١٦ ديسيبل.
- وكانت السعة في وقت مبكر، منتصف، في وقت متأخر وهولو الانقباضي نفخة عند نقطة التسمع قمة في نطاق -٢٤ ديسيبل إلى -٢٢ ديسيبل.

وأظهرت نتائج تحليل سليم الرئة ما يلي:

- كان اتساع كل الأصوات الرئة العادية وغير العادية في حدود - ٥٠ إلى - ٣٠ ديسيبل ديسيبل.
- كان الطيف الترددي من الصوت العادي الرئة الحويصلي ١٢٠ هرتز.
- الطيف تردد الخشنة الخشخشة كان الصوت ٣٣٠ هرتز.
- كان الطيف الترددي للصوت صرير الشهيق ٢٧٥ هرتز.
- كان الطيف الترددي للصوت الاحتكاك الجنبى ٥٨ هرتز.
- كان الطيف الترددي من الصفير الصوت ١٩٨ هرتز.



جامعة الإسكندرية
معهد البحوث الطبية
قسم الفيزياء الحيوية الطبية

تصميم وبناء مجموعة سماعات طبية إلكترونية لتشخيص أمراض القلب والرئة

رسالة مقدمة

بقسم الفيزياء الحيوية الطبية - معهد البحوث الطبية - جامعة الإسكندرية
ضمن متطلبات درجة

الدكتوراه

فى

الفيزياء الحيوية الطبية

من

محمد سيد منصور محمد

بكالوريوس العلوم الطبية التطبيقية، جامعة ٦ أكتوبر، ٢٠٠٣
ماجستير الفيزياء الحيوية الطبية، معهد البحوث الطبية، جامعة الإسكندرية، ٢٠١١



جامعة الإسكندرية
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رسالة مقدمة من

محمد سيد منصور محمد

للحصول على درجة

الدكتوراه فى الفيزياء الحيوية الطبية

موافقون

لجنة المناقشة والحكم على الرسالة

أ.د/ إسماعيل إبراهيم حجازى

أستاذ الكيمياء الحيوية

كلية الطب

جامعة الأزهر بنين

أ.د/ محمد كمال الدين نصره

أستاذ الفيزياء الحيوية الطبية

معهد البحوث الطبية

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أستاذ الفيزياء الحيوية الطبية

كلية العلوم الطبية

جامعة فاروس

أ.د/ سهير الخولى

أستاذ الفيزياء الحيوية الطبية

معهد البحوث الطبية

جامعة الإسكندرية

/ / التاريخ

المشرفون

موافقون

الدكتور / نيفان محمود فكرى

أستاذ بقسم الطبيعة الحيوية الطبية

معهد البحوث الطبية

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الدكتور / مصطفى مصطفى محمد

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الدكتور/ محمد أحمد بندق

مدرس الهندسة الحيوية الطبية

بكلية العلوم الطبية التطبيقية

جامعة ٦ أكتوبر

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