```
typedef double XFLOAT;
    typedef double OTA_FLOAT;
namespace POLQAV2
BOOL stringEndsWithWav (const CNewStdString &s) {
   n = s. GetLength ();
   if (n < 3) {
       return FALSE;
   return TRUE;
        }
   return FALSE;
CTimeSeries::CTimeSeries()
   aInitialized = FALSE;
   aName = "";
   aHeaderDelayInBytes = 0;
   aTrailerDelayInBytes = 0;
   aStereoInFile = FALSE;
BOOL CTimeSeries::Initialize(CNewStdString pName, CPOLQAData *polqaHandle)
    aName = pName;
    if (aInitialized) {
       if (!gBatchMode) {
       return FALSE;
   this->POLQAHandle = polgaHandle;
   statics = polqaHandle->statics;
   ASSERT(polqaHandle->statics->nrTimesSamples > 0);
   SetSize(polqaHandle->statics->nrTimesSamples);
   aInitialized = TRUE;
   SetToConstant(0.);
   return true;
}
void CTimeSeries::SetToConstant (XFLOAT pValue)
   matbSet(pValue, this->m_pData, statics->nrTimesSamples);
void CDoubleArray::RatioOf (const CDoubleArray &pNominator, const CDoubleArray
&pDenominator, XFLOAT pFuzz)
    int i, range;
   int n = GetSize ();
    for (i = 0; i < n; i++) {</pre>
       int j;
       XFLOAT totalWeight = 0;
       range = 30;
```

```
for (j = -range; j <= range; j++) {</pre>
            if ((i + j >= 0) \&\& (i + j < n)) {
                XFLOAT weight = 1.0 / (1.0 + 0.2*(XFLOAT)abs(j));
                this->m_pData[i+j] += weight * (pNominator.m_pData[i+j] + pFuzz) /
(pDenominator.m_pData[i+j] + pFuzz);
                 totalWeight += weight;
        this->m_pData[i] /= totalWeight;
    }
}
void CDoubleArray::CompressOf (const CDoubleArray &pThat, XFLOAT pConstant)
    int n = GetSize ();
    for (i = 0; i < n; i++) {</pre>
        this->m_pData[i] = pow (pThat.m_pData[i], pConstant);
}
XFLOAT CDoubleArray::Power (int pStartIndex, int pStopIndex) const
    int
           i;
    XFLOAT power;
    CNewStdString s;
    power = 0;
    if (pStartIndex < 0)</pre>
        if (!gBatchMode)
        } else {
            s. Format ("TimeSeries. Power : start index negative! " + aName);
            gLogFile. WriteString (s);
        exit (1);
    if (pStartIndex > pStopIndex) {
        if (!gBatchMode) {
        } else {
            s. Format ("TimeSeries. Power : stop index exceeds start index!\n" + aName);
            gLogFile. WriteString (s);
        exit (1);
    int n = GetSize ();
    if (pStopIndex > n) {
        if (!gBatchMode) {
        } else {
            s. Format ("TimeSeries. Power : stop index exceeds length!\n" + aName);
            gLogFile. WriteString (s);
        exit (1);
    for (i = pStartIndex; i < pStopIndex; i++) {</pre>
        XFLOAT h = this->m_pData[i];
        power += h * h;
    power /= (pStopIndex - pStartIndex);
    return power;
}
XFLOAT CDoubleArray::PowerInBand(CPOLQAData *POLQAHandle, XFLOAT pLowerFrequency, XFLOAT
pUpperFrequency) const
    XFLOAT result = 0;
    const XFLOAT frequencyResolutionHz = POLQAHandle->statics->aFrequencyResolutionHz;
    for(int bandIndex = 0; bandIndex < GetSize(); bandIndex++)</pre>
```

```
const XFLOAT frequency = bandIndex * frequencyResolutionHz;
        if((frequency >= pLowerFrequency) && (frequency <= pUpperFrequency))</pre>
            result += this->m_pData[bandIndex];
    return result;
void CDoubleArray::InvDb2 (CPOLQAData *POLQAHandle,
                           XFLOAT
                                     pBasisDb [][2],
                                      pNumberOfPointsBasis,
                            int
                           XFLOAT
                                      pDeltaDb [][2],
                           int
                                      pNumberOfPointsDelta)
{
    XFLOAT
           centreOfBandHz;
    XFLOAT
           gainDb;
    XFLOAT gain;
    const XFLOAT freqRes = POLQAHandle->statics->aFrequencyResolutionHz;
    const int size = GetSize();
    for(int bandIndex = 0; bandIndex < size; bandIndex++)</pre>
        centreOfBandHz = freqRes * (XFLOAT) bandIndex;
        gainDb = interpolate (centreOfBandHz, pBasisDb, pNumberOfPointsBasis);
        gainDb += interpolate (centreOfBandHz, pDeltaDb, pNumberOfPointsDelta);
        gain = pow (10.0, gainDb / 10.0);
        this->m_pData[bandIndex] = gain;
}
void CDoubleArray::TimeAvgOf(const CPOLQAData *POLQAHandle, const CHzSpectrum &pThat)
    XFLOAT result;
           count;
    const int stopFrameIdx = POLQAHandle->statics->stopFrameIdx;
    const int aNumberOfHzBands = POLQAHandle->statics->aNumberOfHzBands;
    for(int bandIndex = 0; bandIndex < aNumberOfHzBands; bandIndex++)</pre>
        result = 0;
        count = 0;
        for(int frameIndex = POLQAHandle->statics->startFrameIdx; frameIndex <=</pre>
stopFrameIdx; frameIndex++)
            result += (pThat.m_pData[frameIndex])[bandIndex];
            count++;
        result /= (XFLOAT) count;
        this->m_pData[bandIndex] = result;
}
const char *CTimeSeries::GetName (void) const
{
    return (const char *) aName;
void CTimeSeries::operator *= (XFLOAT pFactor)
{
    matbMpy1(pFactor, this->m_pData, statics->nrTimesSamples);
void CTimeSeries::operator= (const CTimeSeries &pInputTimeSeries)
{
    matbCopy(pInputTimeSeries.m_pData, this->m_pData, statics->nrTimesSamples);
XFLOAT CTimeSeries::Envelope (CPOLQAData *POLQAHandle, int pStartIndex, int frameLength)
const
    XFLOAT envelope;
```

```
SmartBufferPolqa SB(POLQAHandle, frameLength);
    XFLOAT *temp = SB.Buffer;
    int length;
    if(pStartIndex + frameLength - 1 < statics->nrTimesSamples)
        length = frameLength;
        length = statics->nrTimesSamples - pStartIndex;
    matbSqr2(this->m_pData+pStartIndex, temp, length);
    envelope = matSum(temp, length);
    envelope /= frameLength;
    envelope = sqrt(envelope);
    return envelope;
}
XFLOAT CTimeSeries::WindowedSample(int pFrameIndex, int pSampleIndex, int pWindowSize)
    XFLOAT
             result;
    int
             i;
    ASSERT ((0 <= pSampleIndex) && (pSampleIndex < pWindowSize));
    i = pFrameIndex * pWindowSize/2 + pSampleIndex;
    if (i < 0) {
        return 0;
    if (i >= this->GetSize()) {
        return 0;
    result = statics->frameWindow[pSampleIndex] * this->m_pData[i];
    return result;
}
BOOL CTimeSeries::ReadFromBuffer (XFLOAT* pSamples, long NumberOfSamples)
    matbCopy(pSamples, m_pData, NumberOfSamples);
    aNumberOfSamples = NumberOfSamples;
    return TRUE;
void upperCase (char *outputString, const char *inputString) {
    int i, n;
    n = strlen (inputString);
    for (i = 0; i < n; ++i) {</pre>
        outputString[i] = (char) toupper (inputString[i]);
}
BOOL CTimeSeries::OpenFile (XFLOAT aSampleFrequencyHz,
                            CNewStdString pSoundFilePathName,
                                 &pNumberOfSamples,
                            int
                                    pStereoIfNotWavFile,
                            int
                                    pRightIfStereo)
{
    unsigned int
                      flen, lengthInBytes, fileLength, sampleFrequency, bytesPerSecond;
    int
                      n;
                      numChannels, bitsPerSample;
    short
    short
                      taq;
                      riffId[5], formatId [5], dataId[5];
    char
    CNewFile
                         soundFile;
    CNewStdString
                            s;
    aHeaderDelayInBytes = 0;
    aTrailerDelayInBytes = 0;
    aRightIfStereo = pRightIfStereo;
    if (!soundFile. Open (pSoundFilePathName, "rb")) {
```

```
if (gBatchMode) {
        exit (1);
    } else {
    return FALSE;
if (!stringEndsWithWav (pSoundFilePathName)) {
    aStereoInFile = pStereoIfNotWavFile;
    soundFile. SeekToEnd ();
    if (pStereoIfNotWavFile) {
        aNumberOfSamples = soundFile. GetLength () / 4;
    } else {
        aNumberOfSamples = soundFile. GetLength () / 2;
} else {
    soundFile. Read (riffId, 4);
    riffId [4] = ' \setminus 0';
    upperCase (riffId, riffId);
    if (0 != strcmp (riffId, "RIFF")) {
        if (!gBatchMode) {
        return FALSE;
    }
    soundFile. Read (&fileLength, 4);
    soundFile. Read (riffId, 4);
    riffId[4] = '\0';
    upperCase (riffId, riffId);
    if (0 != strcmp (riffId, "WAVE")) {
        if (!gBatchMode) {
        return FALSE;
    }
    soundFile. Read (formatId, 4);
    formatId [4] = ' \setminus 0';
    soundFile. Read (&flen, 4);
    soundFile. Read (&tag, 2);
    if (tag != 1) {
        if (!gBatchMode) {
        return FALSE;
    }
    soundFile. Read (&numChannels, 2);
    switch (numChannels) {
    case 1:
        aStereoInFile = FALSE;
        break;
    case 2:
        aStereoInFile = TRUE;
        break;
    default:
        if (!gBatchMode) {
        return FALSE;
    }
    soundFile. Read (&sampleFrequency, 4);
    if (aSampleFrequencyHz != (XFLOAT)sampleFrequency)
        return FALSE;
    }
```

```
soundFile. Read (&bytesPerSecond, 4);
        soundFile. Read (&tag, 2);
        soundFile. Read (&bitsPerSample, 2);
        if (bitsPerSample != 16) {
            exit (1);
        }
        soundFile. Read (dataId, 4);
        dataId[4] = ' \setminus 0';
        upperCase (dataId, dataId);
        if (0 != strcmp (dataId, "DATA")) {
            exit (1);
         }
        soundFile. Read (&lengthInBytes, 4);
        aHeaderDelayInBytes = soundFile. GetPosition ();
        n = soundFile. GetLength ();
        aTrailerDelayInBytes = n - aHeaderDelayInBytes - lengthInBytes;
        aNumberOfSamples = n/2 - aHeaderDelayInBytes/2 - aTrailerDelayInBytes/2;
        if (aStereoInFile) {
            aNumberOfSamples /= 2;
    if (!aStereoInFile) {
        aRightIfStereo = FALSE;
    pNumberOfSamples = aNumberOfSamples;
    soundFile. Close ();
    return TRUE;
}
short SwapBytes (short a) {
    short b = (short) ((a & 0xff) << 8);
    short c = (short) ((a & 0xff00) >> 8);
    return (short) (b | c);
BOOL CTimeSeries::ReadFromDisk (const CNewStdString &pSoundFilePathName,
                                 long
                                                      pNumberOfSamples,
                                                      pSwapBytes,
                                 int
                                 XFLOAT*
                                                      pChecksum)
{
    short
                        h;
    int
    short
                         *buffer;
    CNewFile
                         soundFile;
    bool couldOpenSoundFile = false;
    int openTrials = 0;
    const int maxOpenTrials = 20;
    couldOpenSoundFile = soundFile. Open (pSoundFilePathName, "rb");
    while(!couldOpenSoundFile && openTrials < maxOpenTrials)</pre>
    {
        couldOpenSoundFile = soundFile. Open (pSoundFilePathName, "rb");
        openTrials++;
        Sleep(200);
    if (!couldOpenSoundFile) {
        if (gBatchMode)
        {
            exit (1);
        }
```

```
else
        return FALSE;
    buffer = new short [2 * pNumberOfSamples];
    ASSERT (sizeof(short) == 2);
    if (aStereoInFile)
        soundFile.Seek(aHeaderDelayInBytes, SEEK_SET);
        if (sizeof (short) * 2 * pNumberOfSamples != soundFile. Read (buffer, sizeof
(short) * 2 * pNumberOfSamples))
        {
            if (!gBatchMode) {
             }
            exit (1);
        }
    }
else
        soundFile.Seek(aHeaderDelayInBytes, SEEK_SET);
        if (sizeof (short) * pNumberOfSamples != soundFile.Read(buffer, sizeof (short) *
pNumberOfSamples)) {
            if (!gBatchMode)
             exit (1);
        }
    }
    for (i = 0; i < pNumberOfSamples; i++) {</pre>
        if (aStereoInFile) {
            if (aRightIfStereo) {
                 h = buffer [2*i+1];
             } else {
                 h = buffer [2*i];
        } else {
            h = buffer [i];
        if (pSwapBytes) {
            this->m_pData[i] = (XFLOAT) SwapBytes (h);
        } else {
            this->m_pData[i] = (XFLOAT) h;
        }
    delete [] buffer;
    soundFile.Close();
    if (pChecksum)
        XFLOAT sum = 0; for (int i=0; i< pNumberOfSamples; i++) sum+=m_pData[i];</pre>
        *pChecksum = sum;
    return TRUE;
BOOL MakeStereoFile (FILE* pOutputFile,
                      const CTimeSeries &pOriginalTimeSeries,
const CTimeSeries &pDistortedTimeSeries,
                      CPOLQAData *POLQAHandle)
    int
                    i;
```

```
int
                   h;
    short
                   *buffer;
    CNewLogFile
                   outputFile(pOutputFile);
    n = POLQAHandle->statics->nrTimesSamples;
    ASSERT (sizeof(short) == 2);
    buffer = new short [2*n];
    for (i = 0; i < n; i++) {
        h = (int) round (pOriginalTimeSeries.m_pData[i]/2.0);
        if (h < -32767) h = -32767;
        if (h > 32767) h = 32767;
        h = (short) round (h);
        buffer [2*i] = (short) h;
        h = (int) round (pDistortedTimeSeries.m_pData[i]/2.0);
        if (h < -32767) h = -32767;
        if (h > 32767) h = 32767;
        h = (short) round (h);
        buffer [2*i + 1] = (short) h;
    }
        outputFile.Write (buffer, sizeof (short) * 2 * n);
    outputFile. Close ();
    delete [] buffer;
    return TRUE;
}
void CTimeSeries::SetToSine(XFLOAT pAmplitude, const XFLOAT pOmega)
{
    for(int i = 0; i < statics->nrTimesSamples; i++)
       this->m_pData[i] = sin(pOmega * i);
    matbMpy1(pAmplitude, this->m_pData, statics->nrTimesSamples);
void CTimeSeries::FilterWith (CPOLQAData
                                                      *POLQAHandle,
                               const CTimeSeries
                                                     &pInputTimeSeries,
                                                     pTaps,
                              XFLOAT*
                               int
                                                      TapsLength)
    int rc;
    rc = matRunFIRFilter(POLQAHandle->mh, pInputTimeSeries.m_pData, this->m_pData,
statics->nrTimesSamples, pTaps, TapsLength, MAT_FIRDelayComp);
    ASSERT(rc == 0);
void CTimeSeries::FilterWith (CPOLQAData
                                                  *POLQAHandle,
                               BOOL
                                                  pInputFFTAvailable,
                              XFLOAT
                                                  pSampleFrequencyHz,
                              XFLOAT
                                                  pFilterCurve [][2],
                                                  pNumberOfPoints,
                               const CTimeSeries &pInputTimeSeries,
                               CDoubleArray
                                                  &pInputFFT,
                              CDoubleArray
                                                  &pOutputFFT)
    XFLOAT
                          factorDb, factor;
                          overallGainFilter = interpolate ((XFLOAT) 1000, pFilterCurve,
    XFLOAT
pNumberOfPoints);
    long
                          i, powerOf2 = 1, order = 0;
                           *x;
    XFLOAT
    XFLOAT
                          frequencyResolution;
    const int aTimeSeriesLength = statics->nrTimesSamples;
    while (powerOf2 < aTimeSeriesLength)</pre>
    {
        powerOf2 *= 2;
        order++;
    }
```

```
SmartBufferPolqa SB_x(POLQAHandle, powerOf2 + 2);
    x = SB_x.Buffer;
    if (!pInputFFTAvailable)
        matbZero(x, powerOf2 + 2);
        matbCopy(pInputTimeSeries.m_pData, x, aTimeSeriesLength);
        for (i = 0; i < aTimeSeriesLength; i++)</pre>
            if (i < 100) {
                x [i] *= (XFLOAT) i / (XFLOAT) 100;
            if (aTimeSeriesLength - 1 - i < 100) {</pre>
                x [i] *= (XFLOAT) (aTimeSeriesLength - 1 - i) / (XFLOAT) 100;
        }
        matRealFft (POLQAHandle->mh, x, order, MAT_Forw);
        pInputFFT.Initialize("pInputFFT", powerOf2 + 2);
        matbCopy(x, pInputFFT.m_pData, powerOf2 + 2);
    else
    {
        matbCopy(pInputFFT.m_pData, x, powerOf2 + 2);
    frequencyResolution = pSampleFrequencyHz / powerOf2;
    for (i = 0; i <= powerOf2/2; i++) {</pre>
        factorDb = interpolate (i * frequencyResolution, pFilterCurve, pNumberOfPoints)
- overallGainFilter;
        factor = pow (10.0, factorDb / 20.0);
        x [2 * i] *= factor;
        x [2 * i + 1] *= factor;
    pOutputFFT.Initialize("pOutputFFT", powerOf2 + 2);
    matbCopy(x, pOutputFFT.m_pData, powerOf2 + 2);
    matCcsFft (POLQAHandle->mh, x, order, MAT_Inv);
    matbCopy(x, this->m_pData, aTimeSeriesLength);
}
inline void SearchForMaxInRange(const XFLOAT *vec, const int startIdx, const int
stopIdx, XFLOAT *curMax, int *curMaxIdx)
{
    const int searchLen = (((0) > (stopIdx - startIdx)) ? (0) : (stopIdx - startIdx));
    if(searchLen)
        int newMaxIdx = 0;
        XFLOAT newMax = 0.0;
        newMax = matMaxExt(vec + startIdx, searchLen, &newMaxIdx);
        newMaxIdx += startIdx;
        if(newMax > *curMax)
            *curMax = newMax;
            *curMaxIdx = newMaxIdx;
        }
    }
XFLOAT CTimeSeries::ReverberationIndicator( CPOLQAData
                                                                 *POLQAHandle,
                                             const XFLOAT
                                                                  pSampleFrequencyHz,
                                             const CTimeSeries
                                                                  &pInputTimeSeriesOrg,
                                             const CTimeSeries
                                                                 &pInputTimeSeriesDis)
                        aTimeSeriesLength = statics->nrTimesSamples;
    const int
    XFLOAT
                        *varianceETC = 0;
    CNewStdString
                        s;
```

```
const int order = matFFTOrder(aTimeSeriesLength);
const int powerOf2 = 1<<order;</pre>
XFLOAT *x1 = (XFLOAT*)matMalloc((powerOf2 + 2) * sizeof(XFLOAT));
XFLOAT *x2 = (XFLOAT*)matMalloc((powerOf2 + 2) * sizeof(XFLOAT));
XFLOAT levelRef = 0;
XFLOAT levelDeg = 0;
levelRef = matbNormL2(pInputTimeSeriesOrg.m_pData, aTimeSeriesLength);
levelRef *= levelRef;
levelDeg = matbNormL2(pInputTimeSeriesDis.m_pData, aTimeSeriesLength);
levelDeg *= levelDeg;
XFLOAT scalingFactor = sqrt((levelRef+1.0)/(levelDeg+1.0));
matbCopy(pInputTimeSeriesOrg.m_pData, x1, aTimeSeriesLength);
matbMpy4(pInputTimeSeriesDis.m_pData, scalingFactor, x2, aTimeSeriesLength);
matbZero(x1 + aTimeSeriesLength, powerOf2 + 2 - aTimeSeriesLength);
matbZero(x2 + aTimeSeriesLength, powerOf2 + 2 - aTimeSeriesLength);
const int hulp2 = (int)floor(pSampleFrequencyHz/(XFLOAT)250.0);
const int windowLength = 2*hulp2 + 1;
for (int i = hulp2; i < (aTimeSeriesLength-hulp2); i++) {</pre>
    matbAbs2(x1+i-hulp2, temp, hulpLength);
    hulpIn = matSum(temp, hulpLength)/(hulpLength);
    matbAbs2(x2+i-hulp2, temp, hulpLength);
    hulpOut = matSum(temp, hulpLength)/(hulpLength);
    if ( (hulpOut>3*hulpIn) && (hulpIn > 500.0) )
        x2 [i] *= (3*hulpIn/hulpOut);
    else
        if ( (hulpOut>4*hulpIn) && (hulpIn >200.0) )
            x2 [i] *= (4*hulpIn/hulpOut);
        else
        {
            if ( (hulpOut>5*hulpIn) )
                x2 [i] *= (5*hulpIn/hulpOut);
        }
    }
SB_temp.Free();
temp = 0;
levelRef = matbNormL2(x1, aTimeSeriesLength);
levelRef *= levelRef;
levelDeg = matbNormL2(x2, aTimeSeriesLength);
levelDeg *= levelDeg;
scalingFactor = sqrt((levelRef+1.0)/(levelDeg+1.0));
matbMpy1(scalingFactor, x2, aTimeSeriesLength);
matRealFft(POLQAHandle->mh, x1, order, MAT_Forw);
matRealFft(POLQAHandle->mh, x2, order, MAT_Forw);
SmartBufferPolga SB_H_ETC(POLQAHandle, powerOf2 + 2);
XFLOAT *H = SB_H_ETC.Buffer;
const XFLOAT samplesPerHz = (XFLOAT)powerOf2/pSampleFrequencyHz;
matbZero(powerSpectrum1, powerOf2 + 2);
matbZero(powerSpectrum2, powerOf2 + 2);
reverbIndicator = 0.0;
for (i = 80*samplesPerHz; i < (5000*samplesPerHz) / 2; i++) {
    a1 = x1 [2 * i];
    b1 = x1 [2 * i + 1];
    a2 = x2 [2 * i];
    b2 = x2 [2 * i + 1];
```

```
powerSpectrum1 [i] = a1 * a1 + b1 * b1;
        powerSpectrum2 [i] = a2 * a2 + b2 * b2;
        H[2 * i] = ((a2 * a1) + (b2 * b1)) / ((a1 * a1) + (b1 * b1));
        H [2 * i + 1] = ((b2 * a1) - (a2 * b1)) / ((a1 * a1) + (b1 * b1));
    matFree(x1); x1 = 0;
    matFree(x2); x2 = 0;
    SmartBufferPolqa SB_varianceETC(POLQAHandle, powerOf2 + 2);
    varianceETC = SB_varianceETC.Buffer;
    matbZero(varianceETC, powerOf2 + 2);
    matCcsFft(POLQAHandle->mh, H, order, MAT_Inv);
    const int lowerSearchRangeETC = (int)floor(pSampleFrequencyHz/15.0);
    const int upperSearchRangeETC = ((((int)(pSampleFrequencyHz*2)) < (powerOf2)) ?</pre>
((int)(pSampleFrequencyHz*2)) : (powerOf2));
    XFLOAT avgETCtail = 0.0;
    XFLOAT *ETC = H;
    H = 0;
    matbAbs1(ETC + lowerSearchRangeETC, powerOf2/2 - lowerSearchRangeETC);
    matbSqrt1(ETC + lowerSearchRangeETC, powerOf2/2 - lowerSearchRangeETC);
    const int L10start = (((lowerSearchRangeETC) > (powerOf2/8 + 1)) ?
(lowerSearchRangeETC) : (powerOf2/8 + 1));
    const int L10stop = powerOf2/2;
    const int L10len = L10stop - L10start;
    SmartBufferPolqa SB_L10(POLQAHandle, L10len);
    XFLOAT *L10 = SB_L10.Buffer;
    matbPow2(ETC + L10start, 10.0, L10, L10len);
    avgETCtail = matSum(L10, L10len);
    SB_L10.Free();
    L10 = 0;
    avgETCtail /= (powerOf2/5.0);
    avgETCtail = pow(avgETCtail, 0.1);
    matbAdd1(-1.9*avgETCtail, ETC +lowerSearchRangeETC, upperSearchRangeETC
-lowerSearchRangeETC);
    matbThresh1(ETC +lowerSearchRangeETC, upperSearchRangeETC -lowerSearchRangeETC, 0.0,
MAT_LT);
    matbThresh1(ETC +lowerSearchRangeETC, upperSearchRangeETC -lowerSearchRangeETC,
0.015, MAT_GT);
    XFLOAT maxETCtemp = 0;
    int maxNumbertemp = 0;
    int maxNumber1 = 0;
    XFLOAT max1ETC = 0.0;
    for (i = lowerSearchRangeETC; i < powerOf2/2; i++) {</pre>
        if ( ETC [i]>max1ETC && i < (pSampleFrequencyHz*2.0) )</pre>
            max1ETC = ETC [i];
            maxNumber1 = i;
        }
    }
    maxNumber1 -= lowerSearchRangeETC;
    if (maxNumber1 < 0) maxNumber1 = 0;</pre>
    const int hulp3 = (int)floor(pSampleFrequencyHz/10.0);
    int maxNumber2 = 0;
    XFLOAT max2ETC = 0.0;
    for (i = lowerSearchRangeETC; i < (maxNumber1-hulp3); i++) {</pre>
```

```
if ( ETC [i]>max2ETC && i < (pSampleFrequencyHz*2.0) ) {</pre>
            max2ETC = ETC [i];
            maxNumber2 = i;
        }
    for (i = (maxNumber1+hulp3); i < power0f2/2; i++) {
        if ( ETC [i]>max2ETC && i < (pSampleFrequencyHz*2.0) ) {</pre>
            max2ETC = ETC [i];
            maxNumber2 = i;
        }
    maxNumber2 -= lowerSearchRangeETC;
    if (maxNumber2 < 0) maxNumber2 = 0;</pre>
    int hulpNumber1 = maxNumber1;
    int hulpNumber2 = maxNumber2;
    if(hulpNumber2 < hulpNumber1)</pre>
        int hulp = hulpNumber2;
        hulpNumber2 = hulpNumber1;
        hulpNumber1 = hulp;
    int maxNumber3 = 0;
    XFLOAT max3ETC = 0.0;
    for (i = lowerSearchRangeETC; i < (hulpNumber1-hulp3); i++) {</pre>
        if ( ETC [i]>max3ETC && i < (pSampleFrequencyHz*2.0) ) {</pre>
            max3ETC = ETC [i];
            maxNumber3 = i;
        }
    for (i = (hulpNumber1+hulp3); i < (hulpNumber2-hulp2); i++) {</pre>
        if ( ETC [i]>max3ETC && i < (pSampleFrequencyHz*2.0) ) {</pre>
            max3ETC = ETC [i];
            maxNumber3 = i;
        }
    for (i = (hulpNumber2+hulp3); i < powerOf2/2; i++) {</pre>
        if ( ETC [i]>max3ETC && i < (pSampleFrequencyHz*2.0) ) {</pre>
            max3ETC = ETC [i];
            maxNumber3 = i;
        }
    }
    maxNumber3 -= lowerSearchRangeETC;
    if (maxNumber3 < 0) maxNumber3 = 0;</pre>
    avgVarETC = 0.0;
    hulpCount = 0;
    for (i = 200; i < powerOf2 + 2 - 200; i += 50) {</pre>
        ETCWindowed = 0.0;
        for (j = i - 200 + 1; j < i + 200; j++) {
    ETCWindowed += ETC [j] * exp (-0.5*((-(i*1.0) + 1)))</pre>
(j*1.0))/(0.4*200))*((-(i*1.0) + (j*1.0))/(0.4*200)));
        if (((ETCWindowed / 200) - avgETC) > 0.0) varianceETC [(i - 200) / 50] +=
pow(((ETCWindowed / 200) - avgETC), 2);
        else varianceETC [(i - 200) / 50] = 0;
        if ( (i>600) && (varianceETC [(i - 200) / 50]>0.001) ) {
            avgVarETC += varianceETC [(i - 200) / 50];
            hulpCount += 1;
        }
    }
    const XFLOAT normFactor = 8000.f/pSampleFrequencyHz;
    6.0*(max2ETC*maxNumber2*normFactor) + 15.0*(max3ETC*maxNumber3*normFactor);
    return reverbIndicator;
int CTimeSeries::GetLength() const
    return statics->nrTimesSamples - SkipAtStart;
```

}

```
}
int CTimeSeries::GetFrameLength() const
{
    return statics->frameLength;
}
```