

```

//
// Programmer:   Craig Stuart Sapp <craig@ccrma.stanford.edu>
// Creation Date: Sat May 13 12:19:13 PDT 2006
// Last Modified: Sat May 20 15:24:51 PDT 2006 (added parameter control)
// Last Modified: Tue Jun 20 19:42:50 PDT 2006 (features and/or bugs added)
// Last Modified: Sun Jul 9 06:42:47 PDT 2006 (automated scaled power slope)
// Last Modified: Sun May 6 01:48:58 PDT 2007 (upgraded to vamp 1.0)
// Filename:    MzPowerCurve.cpp
// URL:        http://sv.mazurka.org.uk/src/MzPowerCurve.cpp
// Documentation: http://sv.mazurka.org.uk/MzPowerCurve
// Syntax:     ANSI99 C++; vamp 0.9 plugin
//
// Description: Calculate the power of an audio signal as it changes
//              over time.
//
// Defines used in getPluginVersion():
#define P_VER "200607100"
#define P_NAME "MzPowerCurve"

#include "MzPowerCurve.h"
#include "MazurkaWindower.h"

#include <math.h>
#include <stdlib.h>

#define ZEROLOG -120.0

#define FILTER_SYMMETRIC 0
#define FILTER_FORWARD 1
#define FILTER_BACKWARD 2

////////////////////////////////////
//
// Vamp Interface Functions
//
////////////////////////////////////
//
// MzPowerCurve::MzPowerCurve -- class constructor.
//
MzPowerCurve::MzPowerCurve(float samplerate) : MazurkaPlugin(samplerate) {
    mz_windowsum = 1.0;
}

////////////////////////////////////
//
// MzPowerCurve::~MzPowerCurve -- class destructor.
//
MzPowerCurve::~MzPowerCurve() {
    // do nothing
}

////////////////////////////////////
//
// optional polymorphic parameter functions inherited from PluginBase:
//
// Note that the getParameter() and setParameter() polymorphic functions
// are handled in the MazurkaPlugin class.
//
////////////////////////////////////
//
// MzPowerCurve::getParameterDescriptors -- return a list of
// the parameters which can control the plugin.
//
MzPowerCurve::ParameterList MzPowerCurve::getParameterDescriptors(void) const {

    ParameterList pdlist;
    ParameterDescriptor pd;

    // first parameter: The size of the analysis window in milliseconds
    pd.identifier = "windowsize";
    pd.name = "Window size";
    pd.unit = "ms";
    pd.minValue = 10.0;
    pd.maxValue = 10000.0;
    pd.defaultValue = 10.0;
    pd.isQuantized = 0;
    // pd.quantizeStep = 0.0;
    pdlist.push_back(pd);

    // second parameter: The hop size between windows in milliseconds
    pd.identifier = "hopsiz";
    pd.name = "Window hop size";
    pd.unit = "ms";
    pd.minValue = 1.0;
    pd.maxValue = 10000.0;
    pd.defaultValue = 10.0;
    pd.isQuantized = 0;
    // pd.quantizeStep = 0.0;
    pdlist.push_back(pd);

    // third parameter: Windowing method
    pd.identifier = "window";
    pd.name = "Weighting window";
    pd.unit = "";
    pd.minValue = 1.0;
    MazurkaWindower::getWindowList(pd.valueNames);
    pd.maxValue = pd.valueNames.size();
    pd.defaultValue = 1.0;
    pd.isQuantized = 1;
    pd.quantizeStep = 1.0;
    pdlist.push_back(pd);
    pd.valueNames.clear();

    // fourth parameter: Factor for exponential smoothing filter
    pd.identifier = "smoothingfactor";
    pd.name = "Smoothing\n (outputs 2-4)";
    pd.unit = "";
    pd.minValue = -1.0;
    pd.maxValue = 1.0;
    pd.defaultValue = 0.2;
    pd.isQuantized = 0;
    // pd.quantizeStep = 0.0;
    pdlist.push_back(pd);

    // fifth parameter: Filtering method
    pd.identifier = "filtermethod";
    pd.name = "Filter method\n (outputs 2-4)";
    pd.unit = "";
    pd.minValue = 0.0;

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pd.maxValue      = 2.0;
pd.defaultValue = 0.0;
pd.isQuantized   = 1;
pd.quantizeStep  = 1.0;
pd.valueNames.push_back("Symmetric");
pd.valueNames.push_back("Forward");
pd.valueNames.push_back("Reverse");
pdlist.push_back(pd);
pd.valueNames.clear();

/*

// sixth parameter: Cut-off threshold for scaled power slope
pd.identifier    = "cutoffthreshold";
pd.name          = "Cut-off threshold\n (output 4 only)";
pd.unit          = "dB";
pd.minValue      = -100.0;
pd.maxValue      = 10.0;
pd.defaultValue = -40.0;
pd.isQuantized   = 0;
// pd.quantizeStep = 0.0;
pdlist.push_back(pd);

// seventh parameter: The width of the cut-off transition region
pd.identifier    = "cutoffwidth";
pd.name          = "Cut-off width\n (output 4 only)";
pd.unit          = "dB";
pd.minValue      = 1.0;
pd.maxValue      = 100.0;
pd.defaultValue = 20.0;
pd.isQuantized   = 0;
// pd.quantizeStep = 0.0;
pdlist.push_back(pd);

*/

return pdlist;
}

////////////////////////////////////
//
// optional polymorphic functions inherited from Plugin:
//

////////////////////////////////////
//
// MzPowerCurve::getPreferredStepSize --
//

size_t MzPowerCurve::getPreferredStepSize(void) const {
    return size_t(getParameter("hopsiz")*getSrate()/1000.0 + 0.5);
}

////////////////////////////////////
//
// MzPowerCurve::getPreferredBlockSize --
//

size_t MzPowerCurve::getPreferredBlockSize(void) const {
    return size_t(getParameter("windowsize")*getSrate()/1000.0 + 0.5);
}

```

```

////////////////////////////////////
//
// required polymorphic functions inherited from PluginBase:
//

std::string MzPowerCurve::getIdentifier(void) const
    { return "mzpowercurve"; }

std::string MzPowerCurve::getName(void) const
    { return "Power Curve"; }

std::string MzPowerCurve::getDescription(void) const
    { return "Power Curve"; }

std::string MzPowerCurve::getMaker(void) const
    { return "The Mazurka Project"; }

std::string MzPowerCurve::getCopyright(void) const
    { return "2006 Craig Stuart Sapp"; }

int MzPowerCurve::getPluginVersion(void) const {
    const char *v = "@@VampPluginID@" P_NAME "@" P_VER "@" __DATE__ "@@";
    if (v[0] != '@') { std::cerr << v << std::endl; return 0; }
    return atol(P_VER);
}

////////////////////////////////////
//
// required polymorphic functions inherited from Plugin:
//

////////////////////////////////////
//
// MzPowerCurve::getInputDomain -- the host application needs
// to know if it should send either:
//
// TimeDomain      == Time samples from the audio waveform.
// FrequencyDomain == Spectral frequency frames which will arrive
// in an array of interleaved real, imaginary
// values for the complex spectrum (both positive
// and negative frequencies). Zero Hz being the
// first frequency sample and negative frequencies
// at the far end of the array as is usually done.
// Note that frequency data is transmitted from
// the host application as floats. The data will
// be transmitted via the process() function which
// is defined further below.
//

MzPowerCurve::InputDomain MzPowerCurve::getInputDomain(void) const {
    return TimeDomain;
}

////////////////////////////////////
//
// MzPowerCurve::getOutputDescriptors -- return a list describing
// each of the available outputs for the object. OutputList
// is defined in the file vamp-sdk/Plugin.h:
//
// .identifier      == short name of output for computer use. Must not
// contain spaces or punctuation.

```

```

// .name           == long name of output for human use.
// .unit           == the units or basic meaning of the data in the
//                 specified output.
// .hasFixedBinCount == true if each output feature (sample) has the
//                 same dimension.
// .binCount       == when hasFixedBinCount is true, then this is the
//                 number of values in each output feature.
//                 binCount=0 if timestamps are the only features,
//                 and they have no labels.
// .binNames       == optional description of each bin in a feature.
// .hasKnownExtent == true if there is a fixed minimum and maximum
//                 value for the range of the output.
// .minValue       == range minimum if hasKnownExtent is true.
// .maxValue       == range maximum if hasKnownExtent is true.
// .isQuantized    == true if the data values are quantized. Ignored
//                 if binCount is set to zero.
// .quantizeStep   == if isQuantized, then the size of the quantization,
//                 such as 1.0 for integers.
// .sampleType     == Enumeration with three possibilities:
// OD::OneSamplePerStep -- output feature will be aligned with
//                 the beginning time of the input block data.
// OD::FixedSampleRate -- results are evenly spaced according to
//                 .sampleRate (see below).
// OD::VariableSampleRate -- output features have individual timestamps.
// .sampleRate     == samples per second spacing of output features when
//                 sampleType is set toFixedSampleRate.
//                 Ignored if sampleType is set to OneSamplePerStep
//                 since the start time of the input block will be used.
//                 Usually set the sampleRate to 0.0 if VariableSampleRate
//                 is used; otherwise, see vamp-sdk/Plugin.h for what
//                 positive sampleRates would mean.
//

```

```
MzPowerCurve::OutputList MzPowerCurve::getOutputDescriptors(void) const {
```

```
    OutputList    list;
    OutputDescriptor od;
```

```
    // First output channel:
```

```
    od.identifier    = "rawpower";
    od.name          = "Raw Power";
    od.unit          = "dB";
    od.hasFixedBinCount = true;
    od.binCount      = 1;
    od.hasKnownExtents = false;
    // od.minValue    = 0.0;
    // od.maxValue    = 0.0;
    od.isQuantized   = false;
    // od.quantizeStep = 1.0;
    od.sampleType    = OutputDescriptor::VariableSampleRate;
    // od.sampleRate  = 0.0;
    list.push_back(od);
```

```
    // Second output channel: (smoothed data)
```

```
    od.identifier    = "smoothpower";
    od.name          = "Smoothed Power";
    od.unit          = "dB";
    od.hasFixedBinCount = true;
    od.binCount      = 1;
    od.hasKnownExtents = false;
    // od.minValue    = 0.0;
    // od.maxValue    = 0.0;
    od.isQuantized   = false;
    // od.quantizeStep = 1.0;
    od.sampleType    = OutputDescriptor::VariableSampleRate;
```

```
    // od.sampleRate  = 0.0;
    list.push_back(od);
```

```
    // Third output channel: (smoothed data slope)
```

```
    od.identifier    = "smoothpowerslope";
    od.name          = "Smoothed Power Slope";
    od.unit          = "dB slope";
    od.hasFixedBinCount = true;
    od.binCount      = 1;
    od.hasKnownExtents = false;
    // od.minValue    = 0.0;
    // od.maxValue    = 0.0;
    od.isQuantized   = false;
    // od.quantizeStep = 1.0;
    od.sampleType    = OutputDescriptor::VariableSampleRate;
    // od.sampleRate  = 0.0;
    list.push_back(od);
```

```
    // Fourth output channel: (smoothed data slope product)
```

```
    od.identifier    = "powerslopeproduct";
    od.name          = "Scaled Power Slope";
    od.unit          = "dB slope";
    od.hasFixedBinCount = true;
    od.binCount      = 1;
    od.hasKnownExtents = false;
    // od.minValue    = 0.0;
    // od.maxValue    = 0.0;
    od.isQuantized   = false;
    // od.quantizeStep = 1.0;
    od.sampleType    = OutputDescriptor::VariableSampleRate;
    // od.sampleRate  = 0.0;
    list.push_back(od);
```

```
    return list;
```

```
}
```

```
////////////////////////////////////
```

```
//
// MzPowerCurve::initialise -- this function is called once
// before the first call to process().
//
```

```
bool MzPowerCurve::initialise(size_t channels, size_t stepsize,
                               size_t blocksize) {
```

```
    if (channels < getMinChannelCount() || channels > getMaxChannelCount()) {
        return false;
    }
```

```
    // step size and block size should never be zero
    if (stepsize <= 0 || blocksize <= 0) {
        return false;
    }
```

```
    setChannelCount(channels);
    setStepSize(stepsize);
    setBlockSize(blocksize);
```

```
    mz_window.makeWindow(getParameterString("window"), getBlockSize());
```

```
    if (mz_window.getWindowType() == "Rectangular" ||
        mz_window.getWindowType() == "Unknown") {
```

```

    mz_windowsum = 1.0;
} else {
    mz_windowsum = mz_window.getWindowSum() / mz_window.getSize();
}

switch (getParameterInt("filtermethod")) {
    case FILTER_FORWARD:
        mz_filterforward = 1;
        mz_filterbackward = 0;
        break;
    case FILTER_BACKWARD:
        mz_filterforward = 0;
        mz_filterbackward = 1;
        break;
    case FILTER_SYMMETRIC:
    default:
        mz_filterforward = 1;
        mz_filterbackward = 1;
}

rawpower.clear();

return true;
}

////////////////////////////////////
//
// MzPowerCurve::process -- This function is called sequentially on the
//   input data, block by block. After the sequence of blocks has been
//   processed with process(), the function getRemainingFeatures() will
//   be called.
//
// Here is a reference chart for the Feature struct:
//
// .hasTimestamp == If the OutputDescriptor.sampleType is set to
//   VariableSampleRate, then this should be "true".
// .timestamp    == The time at which the feature occurs in the time stream.
// .values       == The float values for the feature. Should match
//   OD:binCount.
// .label        == Text associated with the feature (for time instants).
//
MzPowerCurve::FeatureSet MzPowerCurve::process(AUDIODATA inputbufs,
    Vamp::RealTime timestamp) {

    if (getChannelCount() <= 0) {
        std::cerr << "ERROR: MzPowerCurve::process: "
            << "MzPowerCurve has not been initialized"
            << std::endl;
        return FeatureSet();
    }

    // calculate the raw power for the given input block of audio.
    // Further processing of the raw power data will be done in
    // the getRemainingFeatures() function.

    int i;
    double value;
    double sum = 0.0;

    if (mz_window.getWindowType() == "Unknown" ||
        mz_window.getWindowType() == "Rectangular") {
        // do unweighted power calculation

        for (i=0; i<getBlockSize(); i++) {
            value = inputbufs[0][i];
            sum += value * value;
        }
    } else {
        // do weighted power calculation
        for (i=0; i<getBlockSize(); i++) {
            value = inputbufs[0][i];
            sum += value * value * mz_window[i];
        }
    }

    float power;
    if (sum <= 0.0) {
        power = ZEROLOG;
    } else {
        power = 10.0 * log10(sum/getBlockSize()/mz_windowsum);
    }

    FeatureSet returnFeatures;
    Feature feature;

    // center the location of the power measurement at the
    // middle of the analysis region rather than at the beginning.
    feature.hasTimestamp = true;
    feature.timestamp = timestamp +
        Vamp::RealTime::fromSeconds(0.5 * getBlockSize()/getSrate());

    feature.values.push_back(power);

    // also store the power measurement for later processing in
    // getRemainingFeatures():
    rawpower.push_back(power);

    returnFeatures[0].push_back(feature);

    return returnFeatures;
}

////////////////////////////////////
//
// MzPowerCurve::getRemainingFeatures -- This function is called
//   after the last call to process() on the input data stream has
//   been completed. Features which are non-causal can be calculated
//   at this point. See the comment above the process() function
//   for the format of output Features.
//
MzPowerCurve::FeatureSet MzPowerCurve::getRemainingFeatures(void) {

    int i;
    double filterk = getParameter("smoothingfactor");
    double oneminusk = 1.0 - filterk;
    int size = rawpower.size();
    std::vector<double> smoothpower(size, true);

    // Difference equation for smoothing: y[n] = k * x[n] + (1-k) * y[n-1]

    // do reverse smoothing: time symmetric filtering to remove filter delays
    if (mz_filterbackward && mz_filterforward) {

        // reverse filtering first
        smoothpower[size-1] = rawpower[size-1];

```

```

for (i=size-2; i>=0; i--) {
    smoothpower[i] = filterk*rawpower[i] + oneminusk*smoothpower[i+1];
}

// then forward filtering
for (i=1; i<size; i++) {
    smoothpower[i] = filterk*smoothpower[i] + oneminusk*smoothpower[i-1];
}
} else if (mz_filterbackward) {
    smoothpower[size-1] = rawpower[size-1];
    for (i=size-2; i>=0; i--) {
        smoothpower[i] = filterk * rawpower[i] + oneminusk * smoothpower[i+1];
    }
} else if (mz_filterforward) {
    // do forward smoothing:
    smoothpower[0] = rawpower[0];
    for (i=1; i<size; i++) {
        smoothpower[i] = filterk * rawpower[i] + oneminusk * smoothpower[i-1];
    }
} else {
    smoothpower = rawpower;
}

FeatureSet returnFeatures;
Feature feature;
feature.hasTimestamp = true;

// process output features #2: smoothed power data

double timeinsec;
for (i=0; i<size; i++) {
    timeinsec = (getBlockSize()*0.5 + i * getStepSize())/getSrate();
    feature.timestamp = Vamp::RealTime::fromSeconds(timeinsec);
    feature.values.clear();
    feature.values.push_back(float(smoothpower[i]));
    returnFeatures[1].push_back(feature);
}

// process output features #3 here: smoothed power slope

std::vector<double> smoothslope(size-1, true);
for (i=0; i<size-1; i++) {
    smoothslope[i] = smoothpower[i+1] - smoothpower[i];
    // adding additional 1/2 of the block size to center the peaks
    // at attack points
    timeinsec = (getBlockSize()*0.5 + (i+0.5)*getStepSize())/getSrate();
    feature.timestamp = Vamp::RealTime::fromSeconds(timeinsec);
    feature.values.clear();
    feature.values.push_back(float(smoothslope[i]));
    returnFeatures[2].push_back(feature);
}

// process output features #4 here: scaled smoothed power slope

double mean = getMean(smoothpower);
double sd = getStandardDeviation(smoothpower);

std::vector<double> productslope(size-1, true);
double cutoff = mean - 1.5 * sd;
double width = sd / 2.0;
double scaling;
for (i=0; i<size-1; i++) {
    scaling = (smoothpower[i] - cutoff)/width;
    scaling = 1.0 / (1.0 + pow(2.718281828, -scaling)); //sigmoid function

```

```

    productslope[i] = smoothslope[i] * scaling;
    // adding additional 1/2 of the block size to center the peaks
    // at attack points
    timeinsec = (getBlockSize()*0.5 + (2*i+1)*getStepSize())/(2.0*getSrate());
    feature.timestamp = Vamp::RealTime::fromSeconds(timeinsec);
    feature.values.clear();
    feature.values.push_back(float(productslope[i]));
    returnFeatures[3].push_back(feature);
}

return returnFeatures;
}

////////////////////////////////////
//
// MzPowerCurve::reset -- This function may be called after data processing
// has been started with the process() function. It will be called when
// processing has been interrupted for some reason and the processing
// sequence needs to be restarted (and current analysis output thrown out).
// After this function is called, process() will start at the beginning
// of the input selection as if initialise() had just been called.
// Note, however, that initialise() will NOT be called before processing
// is restarted after a reset().
//
void MzPowerCurve::reset(void) {
    rawpower.clear();
}

////////////////////////////////////
//
// Non-Interface Functions
//
////////////////////////////////////
//
// MzPowerCurve::getMean --
//
double MzPowerCurve::getMean(std::vector<double>& data) {
    double sum;
    int i;

    sum = 0.0;
    for (i=0; i<(int)data.size(); i++) {
        sum += data[i];
    }
    return (sum / data.size());
}

////////////////////////////////////
//
// MzPowerCurve::getStandardDeviation --
//
double MzPowerCurve::getStandardDeviation(std::vector<double>& data) {
    double mean = getMean(data);
    double sum = 0.0;

```

```
double value;  
int i;  
  
for (i=0; i<(int)data.size(); i++) {  
    value = data[i] - mean;  
    sum += value * value;  
}  
  
return sqrt(sum / data.size());  
}
```