

Filterbanks and block-processing in LTFAT

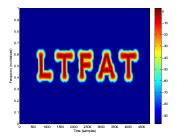
Zdeněk Průša

Acoustics Research Institute (ARI) Austrian Academy of Sciences, Vienna

3rd SPLab Workshop, 1.11.2013



Introduction



is a <u>Matlab/Octave toolbox</u> for working with time-frequency analysis and synthesis. It is intended both as an educational and a computational tool. The toolbox provides a large number of linear transforms including Gabor and wavelet transforms along with routines for constructing windows (filter prototypes) and routines for manipulating coefficients.



- Started in 2004 by Peter L. Søndergaard, 1.0 released in 2011.
- Tested and well documented mat2doc
- MEX/OCT interfaces to the backend lib in C.
- Build system independent of Matlab's mex command.
- Cross-platform, Matlab/Octave, open source, GPL3
- http://ltfat.sourceforge.net



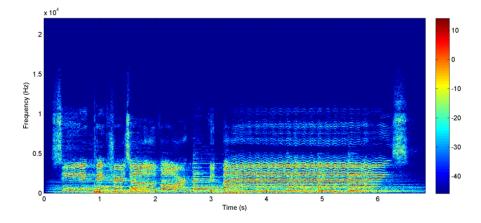
Discrete Gabor Transform R = 16...a picture is worth a thousand words ...

x 10⁴ 2 1.5 -10 Frequency (Hz) -20 -30 0.5 40 6 Time (s)

F = frame('dgtreal',{'hann',882},60,1000);
plotframe(F,frana(F,f),fs,'dynrange',60);



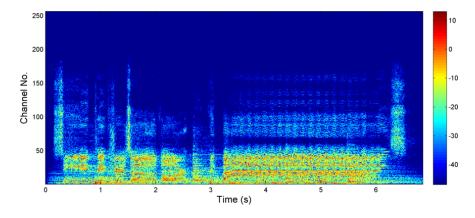
Windowed MDCT R = 1



F = frame('wmdct',{'hann',882},441);



Wavelet Packet subtree R = 1

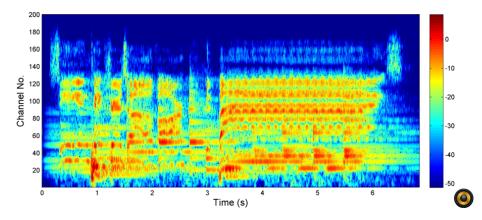


F = frame('wfbt',{'sym10',8})



AR

Erblets $R \sim 12.6$



[g,a]=erbfilters(fs,'fractional','L',numel(f),'M',200,'real')
F = frame('filterbankreal',g,a,numel(g));



Current state of LTFAT

- Filterbanks
- Block-processing framework (and live demonstration)



Current state of LTFAT

Current development version 1.4.2. Version 2 until end of the year!

Main features in LTFAT 2.0:

- Frames framework
- Wavelets module
- Block-processing framework



- The mathematical idea of a "frame" fits well with the notion of class in OOP:
- Each frame has some properties: upper and lower bounds, redundancy, etc.:
 - \implies object attributes.
- Each frame is always associated with analysis and synthesis operators:
 - \implies object methods.
- Simple custom object system using structs.
 - Old (pre 2008a) and new OOP in Matlab.
 - Octave compatibility.



- F = frame create a new freame
- frana(F,...) frame analysis operator
- frsyn(F,...) frame synthesis operator
- framematrix(F,...) matrix form of syntesis operator
- framedual(F,...) construct a dual frame
- frametight(F,...) construct a tight frame
- franalasso(F,...) minimizes $\frac{1}{2}||(f Fc)||_2^2 + \lambda ||c||_1$ (F)ISTA
- franaiter(F,...) iterative analysis using synthesis operator
- frsyniter(F,...) iterative synthesis using analysis opearator
- frsynabs(F,...) synthesis using only abs. values (Griffin-Lim)
- frameaccel(F,L) precompute stuff for given length
- plotframe(F,...) plot frame coefficients

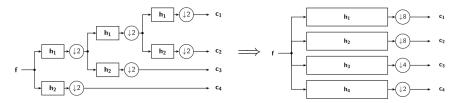


Wavelets module

- fwt Discrete Wavelet Transform (Mallat's algorithm)
- ufwt Undecimated fwt (À-trous algorithm).
- wfbt/uwfbt (Undecimated) Arbitrary tree-shaped Wavelet filterbank.
- wpfbt/uwpfbt (Undecimated) Arbitrary tree-shaped Wavelet filterbank.
- wpbest Best basis selection from bases derived from the wavelet packet.
- fwt2 Basic 2D Discrete wavelet transform.
- plotwavelets common plotting routine.
- Wavelet filters library.
- Helper functions for building FB trees.



- Arbitrary number of filters in the basic filterbank framelets, etc.
- Arbitrary filter trees DT-ℂWT
- fwt2filterbank, wfbt2filterbank tree filterbank conversion routines using multirate identity.





Common routines for FIR, frequency defined and band-limited filters.

$$c_m(n) = \sum_{l=0}^{L-1} f(l) g_m(a_m n - l), \qquad (1)$$

where $L = k \cdot \text{lcm}(a_m)$, $k \in \mathbb{Z}^+$, $f \in \mathbb{C}^L$ and $a_m n - l$ is computed modulo L.

$$\hat{f}(l) = \sum_{m=0}^{M-1} \sum_{n=0}^{L/a_m - 1} c_m(n) \,\tilde{g}_m(l - a_m n) \,, \tag{2}$$

Filter generating routines:

- firfilter struct, main fields .h, .offset
- blfilter struct, main fields .H, .foff

Effective implementation in C.



Filterbanks (u)filterbank/ifilterbank

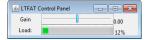
Two purposes:

- A computational routine.
- Filterbank itself as a Frame.
- filterbankdual, filterbankbounds dual filterbanks and frame bounds for <u>uniform</u> and <u>painless</u> filterbanks.
- nonu2ufilterbank nonuniform to uniform filterbank transform. Each filter g_m is replaced by $p = lcm(a_m)/a_m$ delayed versions of itself $z^{-ka_m}G_m(z)$ for k = 0, ..., p - 1



A simple framework for a real-time audio processing directly from Matlab/Octave.

```
block('playrec');
p = blockpanel({'GdB','Gain',-20,20,0,21});
while p.flag
  gain = blockpanelget(p,'GdB');
  f = blockread(1024);
   blockplay(f*10^(gain/20));
end
p.close();
```





```
Based on:

Portaudio (http://www.portaudio.com) and

Playrec (http://www.playrec.co.uk).
```

Main features:

- Interfaces to JACK, ASIO, etc., channel patching.
- No additional toolbox dependency.

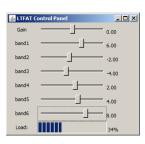
Limitations:

- At 44,1 kHz, block sizes \sim 1000 samples \implies latency \sim 23ms.
- Inherent latency issues from Portaudio.

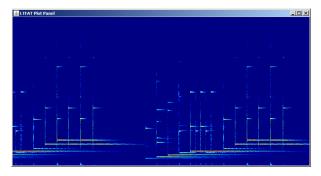


Block-processing GUI

Configurable control panel



Real-Time visualization



JAVA based, independent of Matlab GUI framework.



Basic idea: Analyze (and synthetize) a block stream by any transform available in the Frames framework.

Two issues:

- Speed backend in C, precomputing using blockframeaccel
- Block artifacts
 - Slicing window
 - Overlap-save/ovelap-add



Half-length block overlapping and weighing by a slicing window to reduce time aliasing.

Advantages:

- Works for any transform.
- Delay depends on the block length and is independent of the transform.
- Slicing windows need not add up to 1 dual slicing window.

Disadvantages:

- Coefficients reflects the shape of the slicing window.
- The blocking artifact can still be perceived.



Employs overlap-save method for the analysis and overlap-add method for the synthesis.

Advantages:

- Coefficients can be processed or visualized directly.
- Completely avoids the blocking artifact.

Disadvantages:

- Requires FIR filters/windows.
- Increased processing delay roughly equal to the longest filter/window length.



Live demo



- Releasing LTFAT 2.0
- Various interfaces to LTFAT or LTFAT backend.
 - S_TOOLS-ST^x acoustic speech and signal processing application developed at ARI.
 - Sonic Visualizer (http://www.sonicvisualiser.org/) open-source audio visualizing and annotating application.
 - Python bindings
- Better GUI for the frame multiplier editor mulaclab.



Thank you for listening.

http://ltfat.sourceforge.net/

Download, try, learn, share, contribute...