A Tetrahedral Microphone Processor for Ambisonic Recording

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Overview



- A short intro to Ambisonics.
- The tetrahedral microphone.
- The Tetraproc application.
- Tetrahedral microphone calibration.
- The Fourmic JACK backend.

A short intro to Ambisonics (1)



- Ambisonics can be viewed on many levels:
 - A systematic way to analyse and represent the spatial structure of a wave field.
 - A systematic way to analyse and represent the directional information in sound.
 - A surround sound technology, including recording and reproduction systems.
- Developed 30 years ago by Brithish mathematicians Michael Gerzon, Peter Craven, e.a.
- Firmly based on physics, maths, and psycho-acoustics.
- Full potential could not be realised at the time.
- Today used 'inside' many professional 5.1 and 7.1 applications.
- Much research (Higher Order Ambisonics) in recent years, mainly in France.

The Oxford University Tape Recording Society (1968)





L to R: Michael Gerzon, Paul Hodges, Peter Craven, Stephen Thornton Picture by P. Hodges, used with permission

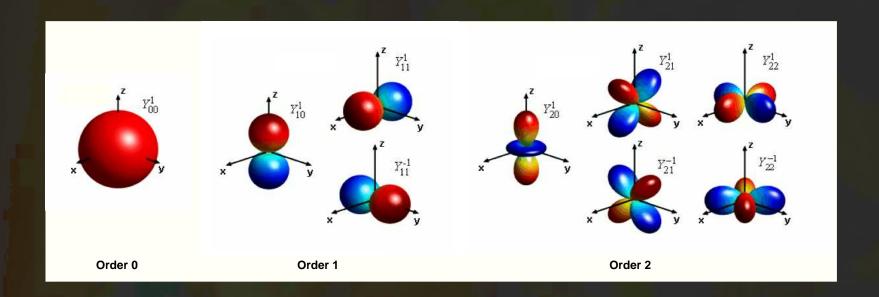
A short intro to Ambisonics (2)



- Ambisonics encodes a sound field using a representation based on Spherical Harmonics.
 - Spherical harmonics represent the 'spatial spectrum' of a sound field.
 - Similar to Fourier analysis, but on the surface of a sphere.
- Given the 'spatial spectrum', a sound field can be reconstructed.
- There are 2M + 1 spherical harmonics of order M.
- At low and medium frequencies, only the lower orders are required.
- Good 3-D results can already be achieved using only orders 0 and 1 (4 signals).
- 2nd order horizontal only systems (5 signals) are far superior to ITU 5.1.
- Today we have working reproduction and recording systems up to 4th order, full 3-D

A short intro to Ambisonics (3)

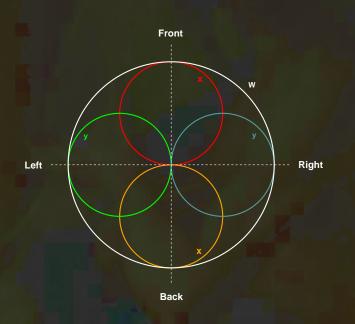




- Spherical harmonics can be seen as polar patterns.
- Low order SH correspond to familiar physical quantities.
 - Order 0: pressure (W).
 - Order 1: the three components of the velocity vector (X,Y,Z).
- ullet A set of signals corresponding to all SH up to order M is called B-format of order M. This is the standard Ambisonics format.

Native B-format recording



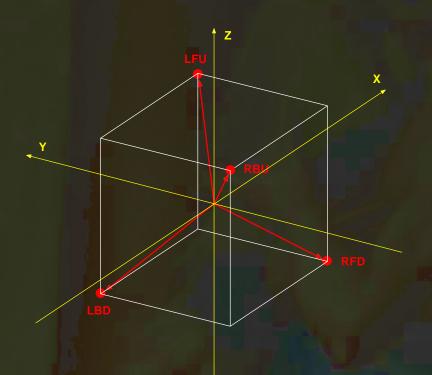




- First-order B-format signals correspond to 'real' microphones.
- An array of 3 mics can be used for 1st order horizontal recording.
- Practical solution for 3-D is a *tetrahedral* microphone.

Tetrahedral Microphones (1)











- Four near-cardiod mics at the vertices of a regular tetrahedron, pointing outwards.
- The four mics shoul ideally be as close together as possible. Practical radius is 15...18 mm.
- ullet The set of four signals (LFU,RFD,LBD,RBU) is known as A-format.

Tetrahedral Microphones (2)



• A to B format conversion is simple (in theory):

$$W = F_0(\omega) \times (LFU + RFD + LBD + RBU)$$

$$X = F_1(\omega) \times (LFU + RFD - LBD - RBU)$$

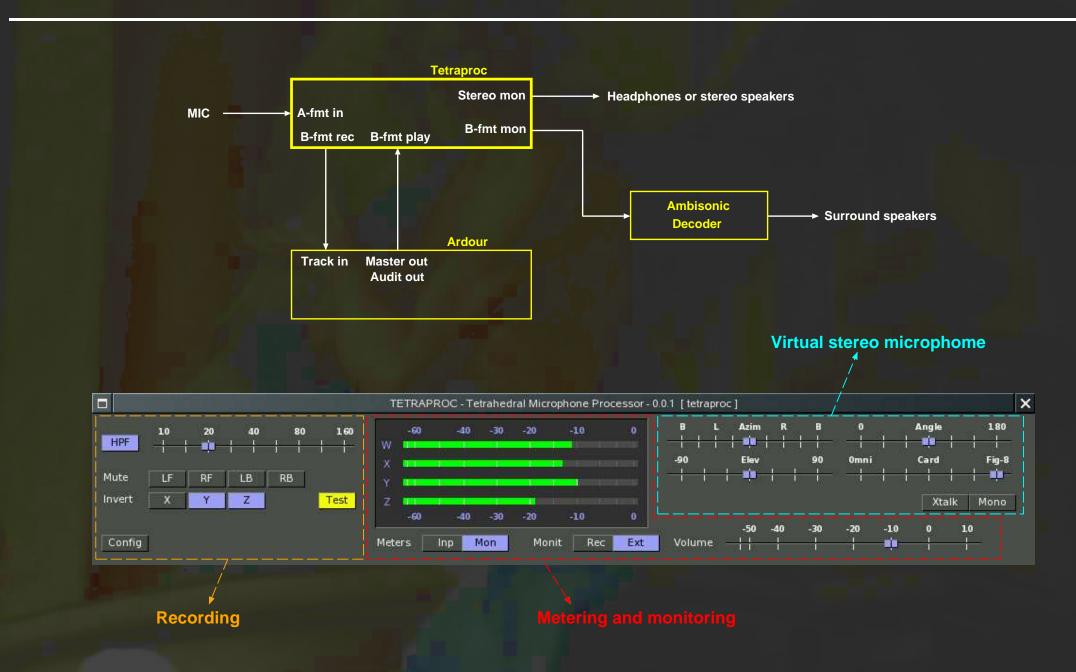
$$Y = F_1(\omega) \times (LFU - RFD + LBD - RBU)$$

$$Z = F_1(\omega) \times (LFU - RFD - LBD + RBU)$$

- The post-matrix filters $F_0(\omega)$ and $F_1(\omega)$ are necessary because the four A-format mics are not really coincident.
- The filters can't be perfect for all directions. They are always a compromise.
- In reality the matrix must be adjusted to compensate for microphone mismatch.
- A practical processor needs some auxiliary functions: high pass filters, test controls, metering, monitoring.

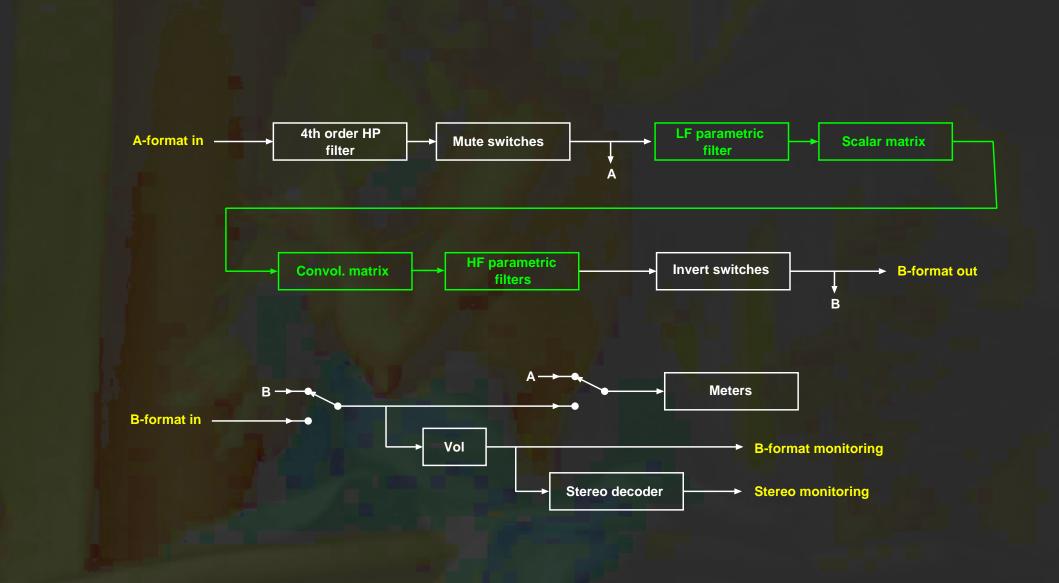
Tetraproc connections and GUI





Tetraproc audio processing (1)





Tetraproc audio processing (2)



• LF parametric EQ.

- Settings derived from measured response.
- Optionally subjective adjustment.
- Scalar A-B matrix.
 - Also corrects gain and directivity mismatch.
 - Used only if no impulse responses are available.
- 4 by 4 convolution matrix, combines:
 - A to B-format conversion.
 - Individual microphone frequency response correction.
 - Frequency-dependent gain and directivity correction.
 - Optionally also the post-matrix parametric EQ.
- Post-matrix parametric EQ.
 - Required for correct encoding at high frequencies.
 - Function can be absorbed into convolution matrix.

Calibration (1)



- Hardware A-B processors are factory aligned for one microphone serial number.
- The same is required for a software implementation.
- The calibration requires some measurements. For each of the four A-format mics:
 - Eight impulse responses at 45 degree intervals in the horizontal plane. These are used to compute the A-B matrix (scalar or convolution), and the postmatrix equalisation.
 - (Optional) On-axis, far-field impulse response, used for individual mic equalisation.
 - (Optional) On-axis, near-field impulse or frequency response, used for low frequency equalisation.

Calibration (2)



- \bullet For an absolutely minimal calibration, the impulse responses can be replaced by 4×8 level measurements, using bandlimited noise in the 500 Hz range.
- Measurements can be done in a large room, or in open air, using e.g. Aliki.
- An omnidirectional reference microphone is required.
- A separate program is used to derive matched processing parameters, and to create a configuration file for Tetraproc.

Calibration (3)



- Calibrating a tetrahedral mic is not a simple procedure...
 - There are many options and variations.
 - Very few people have any real experience with this problem.
 - Performing acoustic measurements correctly can be tricky.
- The methods used are based on procedures recommended by Soundfield microphone designer Richard Lee, together with new possibilities offered by IR measurement software.
- More work will be required to make the calibration as easy as possible for a non-expert user.

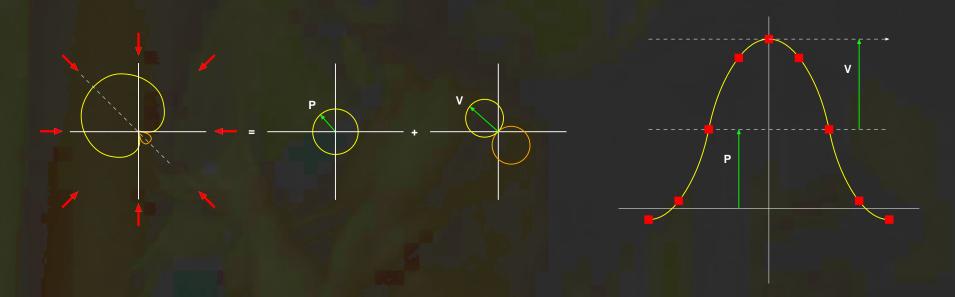
Calibration - Microphone response correction



- MF and HF equalisation using far-field, on-axis impulse response.
 - Uses only the first 5...20 milliseconds of the IR, no LF equalisation.
 - Kirkeby inversion of IR, with manual correction.
 - Result used in convolution matrix.
- LF equalisation based on near-field measurement using a small source.
 - Corrected for near-field effects using known distance and directivity.
 - Used to set the LF parametric equaliser.
- These procedures do not compensate for directivity mismatch.

Calibration - Gain and directivity mismatch (1)





- Small errors in sensitivity and directivity of the A-format mics have a big impact on the first order B-format signals.
- Eight measurements in the horizontal plane allow to compute a cross section of the polar diagram for each of the four A-format microphones.
- ullet P and V can be computed from 2 values, but using all 8 is much more accurate.
- The real V is $\sqrt{3/2}$ times the value in the horizontal plane.

Calibration - Gain and directivity mismatch (2)



Given P and V for each mic, a matched A-B transformation matrix can be computed. If (x, y, z) are the direction cosines, the ouputs of the four mics can be expressed as:

$$LF(x, y, z) = P_{LF} + V_{LF} \times (x + y + z)$$

 $RF(x, y, z) = P_{RF} + V_{RF} \times (x - y - z)$
 $LB(x, y, z) = P_{LB} + V_{LB} \times (-x + y - z)$
 $RB(x, y, z) = P_{RB} + V_{RB} \times (-x - y + z)$

The required B-format signals are:

$$W(x, y, z) = 1$$

$$X(x, y, z) = x$$

$$Y(x, y, z) = y$$

$$Z(x, y, z) = z$$

Expressing (W, X, Y, Z) as linear combinations of (LF, RF, LB, RB) now only requires solving sets of linear equations.

Calibration - Post-matrix filtering (1)



- A complex matter. Result is a compromise and must be evaluated subjectively.
- ullet Automatic derivation of the post-matrix filters using the measured P,V values over the entire frequency range is a research project...
 - The maths get complicated when delays are taken into account.
 - A simple first order microphone model is no longer valid.
 - Complex diffraction effects make the analysis almost impossible.

Calibration - Post-matrix filtering (2)



- Good results seem to be achieved by equalising the cardinal directions.
 - Process the IR using the computed mic EQ and A-B matrix.
 - Adjust parametric EQ using displayed frequency and phase plots.
 - Or use an automated IR inversion procedure, with manual correction.
- Alternatively, equalise the diffuse field response (average power over all directions). This requires more measurements.

Calibration - Post-matrix filtering (3)



- If no IR measaurements are available: use a 'standard' filter.
- Theoretical filters depend on the radius and the directivity.
- There is very little available literature, and no agreement.
- The following are all quite different:
 - The theoretical diffuse field response.
 - The theoretical cardinal direction response.
 - The filter implementation proposed by Gerzon.
 - The filters actually used in HW processors.
- More experimentation will be required.

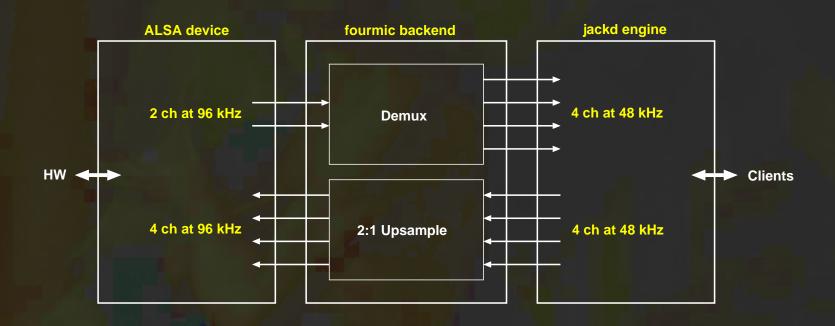
The Fourmic backend (1)



- Using A-format mics requires capture channels with precisely matched gain controls.
- Core Sound's 4Mic interface multiplexes 4 channels on a stereo SPDIF stream at twice the sample frequency.
- Where to do the demultiplexing?
 - In the ALSA drivers.
 - In the user space ALSA library.
 - In an ALSA library plugin.
 - In the application.
 - $-\dots$
 - In the JACK backend.

The Fourmic backend (2)





- The fourmic backend plugin provides:
 - Demultiplexing of the 2-channel stream to the four A-format signals.
 - Upsampling of up to four playback channels for monitoring.
- Demuxing introduces an ambiguity. Automatic resolution using the 4Mic test signal may be possible.

Tetraproc — Things to do



Lots of work ...

- Manual and documentation.
- Improve the calibration procedure.
- Make some recordings.
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Acknowledgements



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