

A **cepstrum** ([/'kɛpstrəm' 'sɛpstrəm'/](#)) is the result of taking the [Inverse Fourier transform](#) (IFT) of the [logarithm](#) of the estimated [spectrum](#) of a signal. It may be pronounced in the two ways given, the second having the advantage of avoiding confusion with 'kepstrum' which also exists (see below). There is a [complex](#) cepstrum, a [real](#) cepstrum, a power cepstrum, and phase cepstrum. The power cepstrum in particular finds applications in the analysis of human speech.

The name "cepstrum" was derived by reversing the first four letters of "spectrum". Operations on cepstra are labelled *quefrequency analysis*, *liftering*, or *cepstral analysis*.

Origin and definition

The *power cepstrum* was defined in a 1963 paper by Bogert et al.[\[1\]](#) The power cepstrum of a signal is defined as the squared magnitude of the inverse Fourier transform of the logarithm of the squared magnitude of the Fourier transform of a signal.[\[2\]](#)

$$\text{power cepstrum of signal} = \left| \mathcal{F}^{-1} \left\{ \log \left(\left| \mathcal{F} \{ f(t) \} \right|^2 \right) \right\} \right|^2$$

A short-time cepstrum analysis was proposed by [Schroeder](#) and Noll for application to pitch determination of human speech.[\[3\]\[4\]\[5\]](#)

The *complex cepstrum* was defined by Oppenheim in his development of homomorphic system theory[\[6\]](#) and is defined as the Inverse Fourier transform of the logarithm (with unwrapped phase) of the Fourier transform of the signal. This is sometimes called the spectrum of a spectrum.

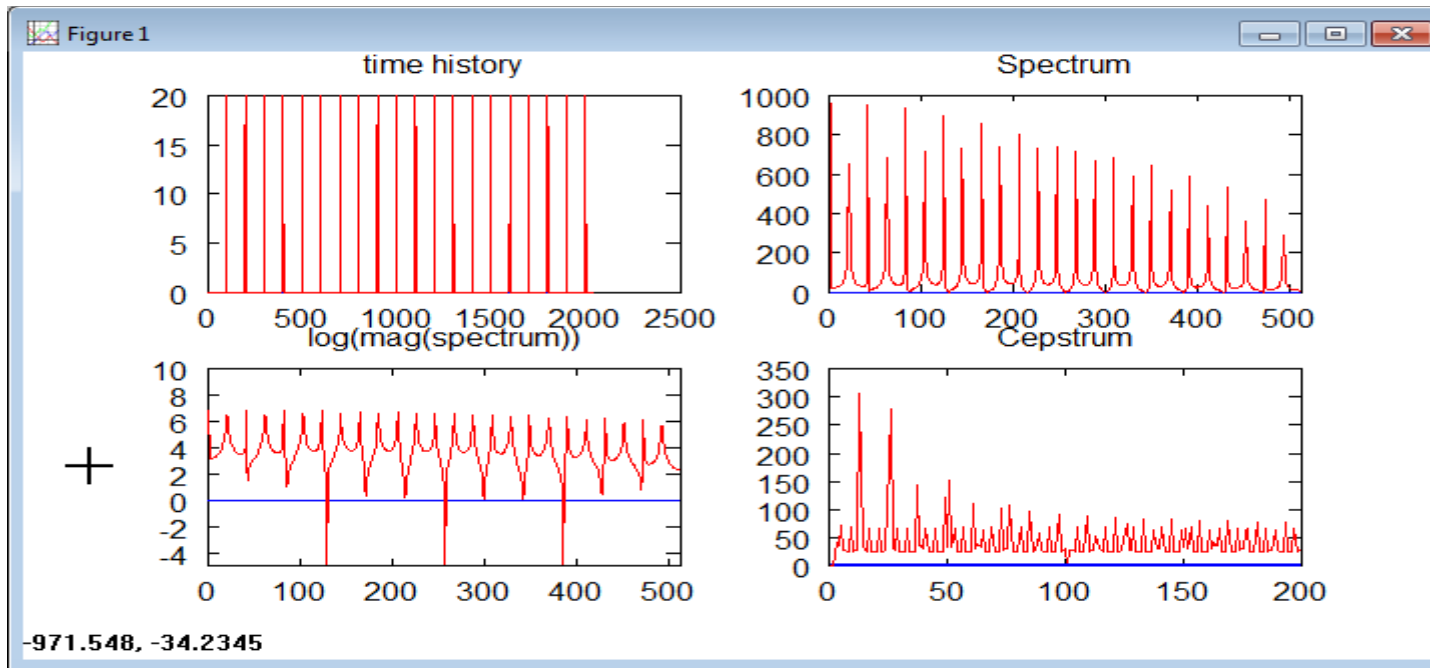
$$\text{complex cepstrum of signal} = \text{IFT}(\log(\text{FT}(\text{the signal})) + j2\pi m) \text{ (where } m \text{ is the integer required to properly } \text{unwrap} \text{ the angle or } \text{imaginary part} \text{ of the complex log function)}$$

The *real cepstrum* uses the [logarithm function](#) defined for real values. The real cepstrum is related to the power via the relationship $(4 * \text{real cepstrum})^2 = \text{power cepstrum}$, and is related to the complex cepstrum as $\text{real cepstrum} = 0.5 * (\text{complex cepstrum} + \text{time reversal of complex cepstrum})$.

Convolution

A very important property of the cepstral domain is that the [convolution](#) of two signals can be expressed as the addition of their complex cepstra:

$$x_1 * x_2 \rightarrow x'_1 + x'_2$$



Steps in forming cepstrum from time history

The complex cepstrum uses the [complex logarithm function](#) defined for complex values. The phase cepstrum is related to the complex cepstrum as $\text{phase spectrum} = (\text{complex cepstrum} - \text{time reversal of complex cepstrum})^2$

The complex cepstrum holds information about magnitude and phase of the initial spectrum, allowing the reconstruction of the signal. The real cepstrum uses only the information of the magnitude of the spectrum.

Many texts define the process as $\text{FT} \rightarrow \text{abs}() \rightarrow \text{log} \rightarrow \text{IFT}$, i.e., that the cepstrum is the "inverse Fourier transform of the log-magnitude Fourier spectrum".[\[7\]](#)[\[8\]](#)

The *kepstrum*, which stands for "Kolmogorov equation power series time response", is similar to the cepstrum and has the same relation to it as expected value has to statistical average, i.e. cepstrum is the empirically measured quantity while kepstrum is the theoretical quantity.[\[9\]](#)[\[10\]](#)

Applications

The cepstrum can be seen as information about rate of change in the different spectrum bands. It was originally invented for characterizing the seismic [echoes](#) resulting from [earthquakes](#) and [bomb](#) explosions. It has also been used to determine the fundamental frequency of human speech and to analyze [radar](#) signal returns. Cepstrum pitch determination is particularly effective because the effects of the vocal excitation (pitch) and [vocal tract](#) (formants) are additive in the logarithm of the power spectrum and thus clearly separate.[\[5\]](#)

The autocepstrum is defined as the cepstrum of the [autocorrelation](#). The autocepstrum is more accurate than the cepstrum in the analysis of data with echoes.

The cepstrum is a representation used in [homomorphic signal processing](#), to convert signals (such as a source and filter) combined by [convolution](#) into sums of their cepstra, for linear separation. In particular, the power cepstrum is often used as a feature vector for representing the human voice and musical signals. For these applications, the spectrum is usually first transformed using the [mel scale](#). The result is called the [mel-frequency cepstrum](#) or MFC (its coefficients are called mel-frequency cepstral coefficients, or MFCCs). It is used for voice identification, [pitch detection](#) and much more. The cepstrum is useful in these applications because the low-frequency periodic excitation from the [vocal cords](#) and the [formant](#) filtering of the [vocal tract](#), which convolve in the [time domain](#) and multiply in the [frequency domain](#), are additive and in different regions in the quefrequency domain.

Cepstral concepts

The [independent variable](#) of a cepstral graph is called the **quefrequency**. The quefrequency is a measure of time, though not in the sense of a signal in the [time domain](#). For example, if the sampling rate of an audio signal is 44100 Hz and there is a large peak in the cepstrum whose quefrequency is 100 samples, the peak indicates the presence of a pitch that is $44100/100 = 441$ Hz. This peak occurs in the cepstrum because the harmonics in the spectrum are periodic, and the period corresponds to the pitch. Note that a pure sine wave should not be used to test the cepstrum for its pitch determination from quefrequency as a pure sine wave does not contain any harmonics. Rather, a test signal containing harmonics should be used (such as the sum of at least two sines where the second sine is some harmonic (multiple) of the first sine).

Liftering

Playing further on the anagram theme, a filter that operates on a cepstrum might be called a *lifter*. A low pass lifter is similar to a low pass filter in the [frequency domain](#). It can be implemented by multiplying by a window in the quefrequency domain and when converted back to the frequency domain, resulting in a smoother signal.