

Time and Moveout Filtering

- Frequency filtering
- Wavelet shaping (deconvolution)
- Dip and Moveout (2-D) filtering
 - ◆ $f-k$ (frequency-wavenumber)
 - ◆ $\tau-p$ (slant stack)
- Reading:
 - › Sheriff and Geldart, Sections 9.5, 9.9, 9.11

Single-channel Filtering

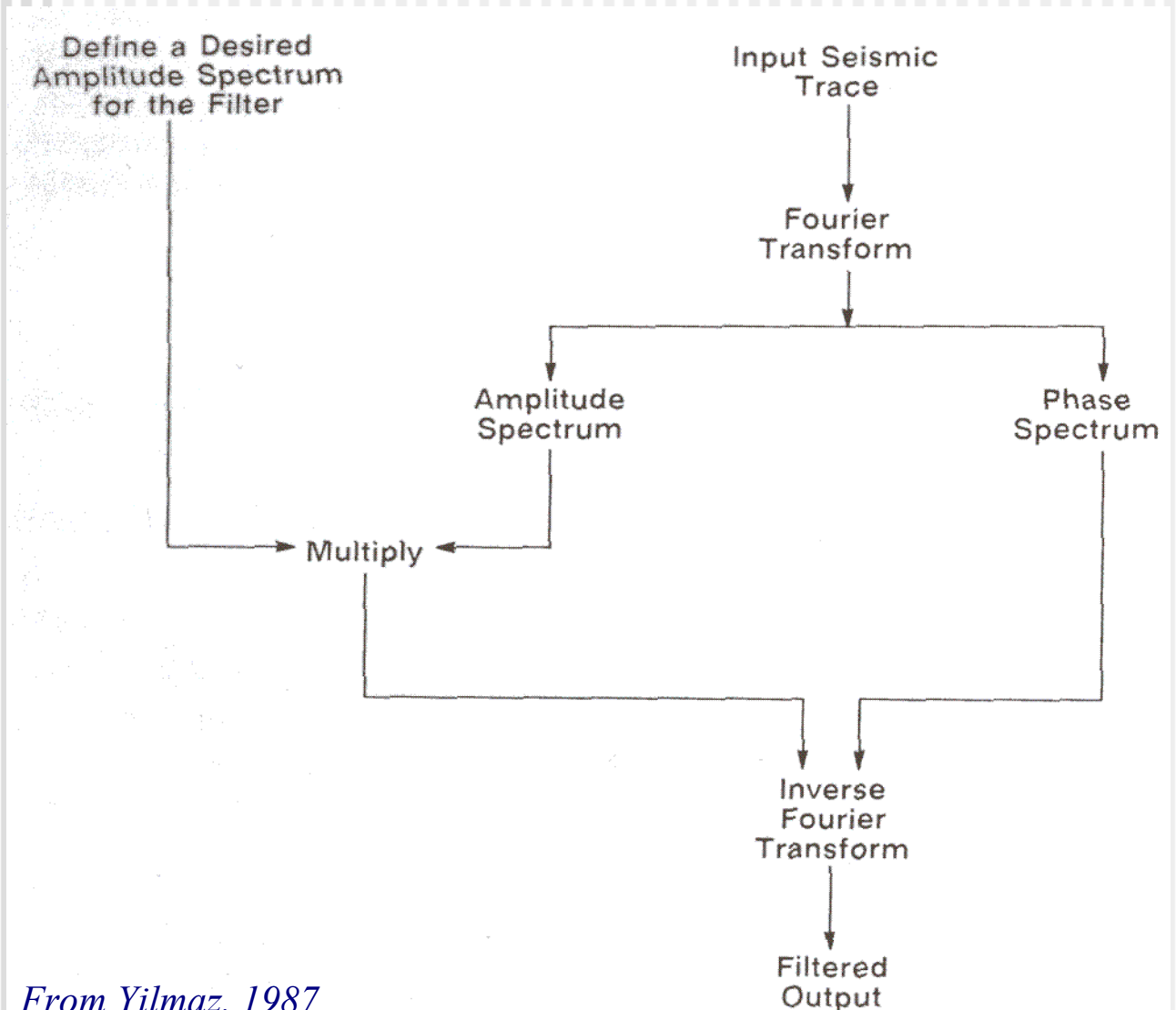
Objectives

- Performed in order to increase the Signal/Noise ratio or to improve signal shape:
 - Modify the frequency band
 - Flatten (“whiten”) the spectrum
 - Convert the wavelet into minimum- or zero-phase (*wavelet shaping*)
 - Minimum-phase wavelet is causal;
 - Zero-phase is better for display and interpretation
 - Normalize the effects of different sensors by bringing them to a common response (*matching filters*)
 - Remove reverberations (*deconvolution*)
- The Filter is always a time series *convolved* with the signal
 - This can always be done in *time* or *frequency* domain

Frequency filtering

Frequency-domain

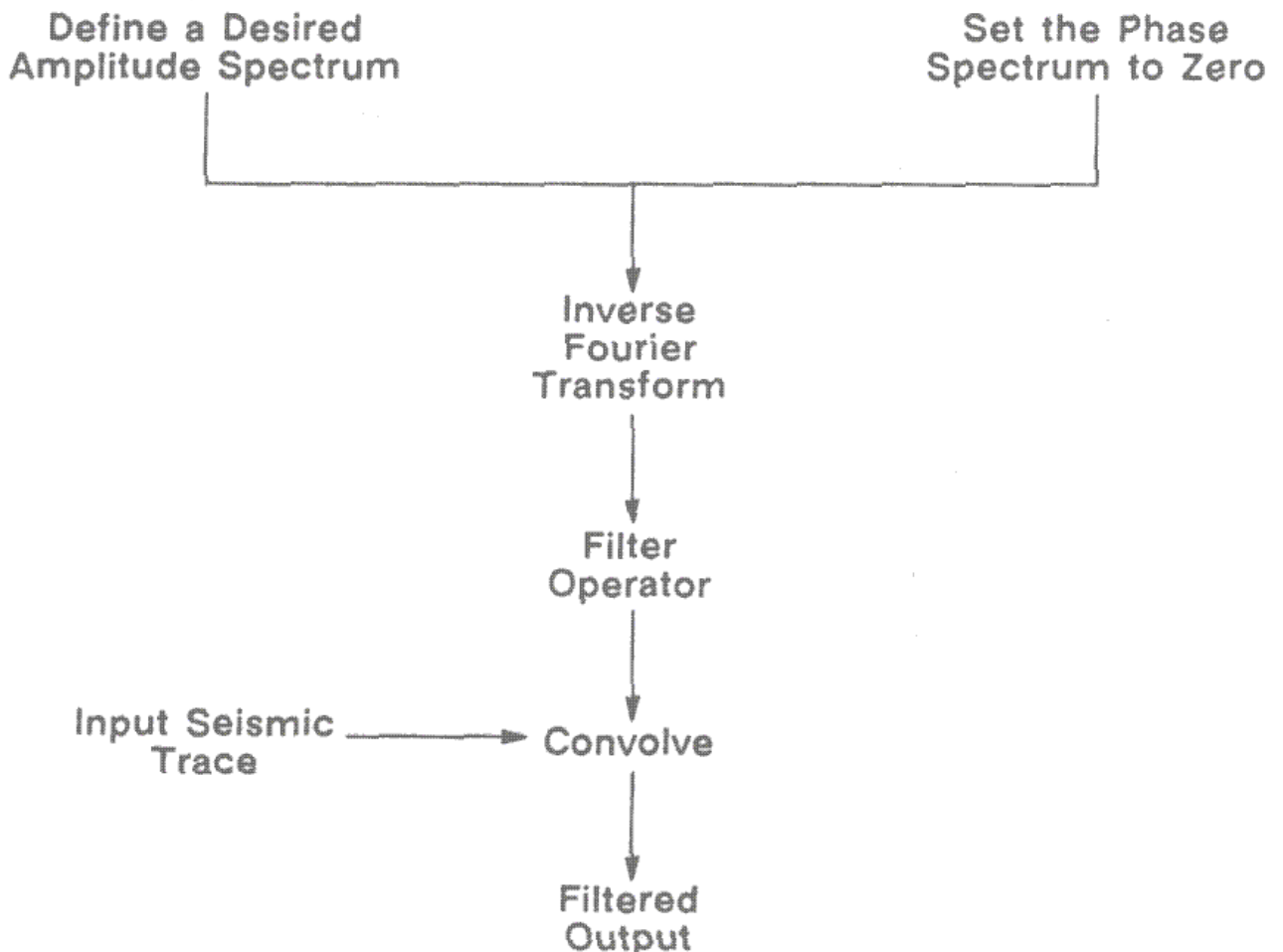
- Most common
- Zero phase filter in order to preserve phase character



Frequency filtering

Time-domain

- This is used only for broad-band (short in time) filters when time-domain convolution is more efficient than forward and inverse FFT



From Yilmaz, 1987

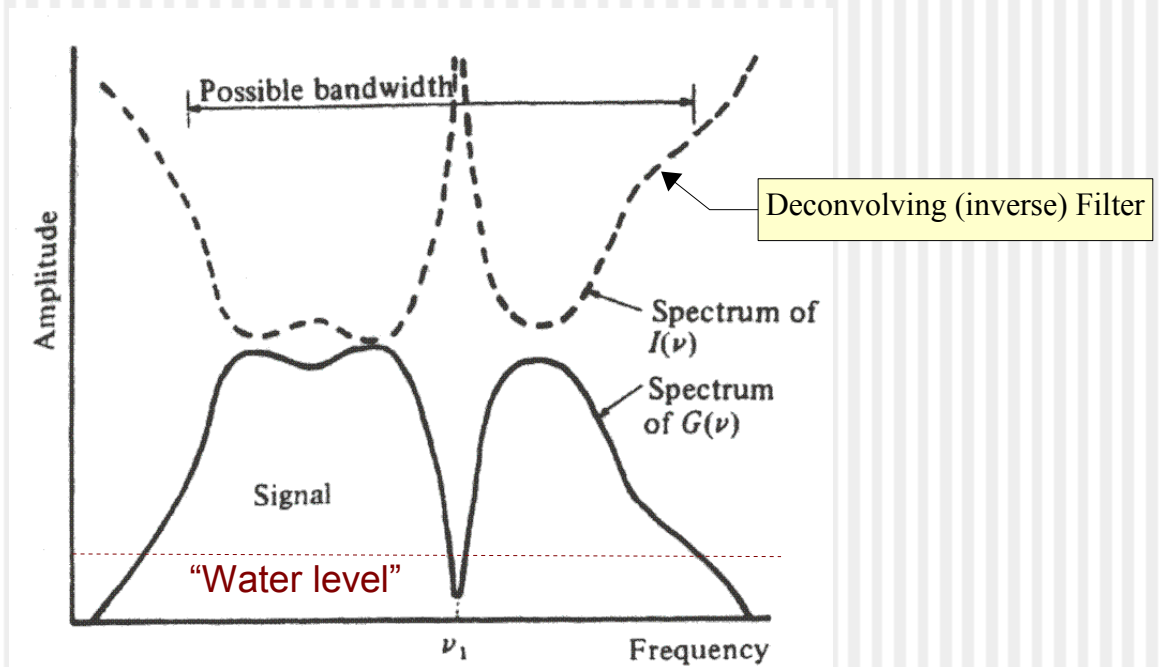
Deconvolution

- Time domain:
 - Changing the shape of the signal to some “desired” waveform
 - Spiking (to a spike)
 - Shaping (to a band-limited pulse).
 - Removal of short-period multiples
 - Prediction-error deconvolution.
- Frequency domain:
 - Flattening the spectrum
 - Spectral broadening, more spiky signal
 - Time-Variant Spectral Whitening (compensates attenuation)
 - Transformation to a zero-phase (symmetric) wavelet.

Deconvolution

Spectral whitening

- Frequency-domain
- The zero-phase inverse filter is constructed of the inverse of signal amplitude. "Spectral holes" corrected by adding 1-2% "pre-whitening" or "water level"



$$A_{inverse}(f) = \frac{1}{A(f) + A_{prewhitening}}$$

$$A_{inverse}(f) = \frac{1}{\max\{A(f), A_{water\ level}\}}$$

Deconvolution

Wiener (least squares)

- Time-domain
- Changes the shape of the signal into some "desired" waveform:

$$u_i^{desired} = \sum_k f_k u_{i-k}$$

- This is solved for f_k by using the Least-Squares method:

$$\sum (u_i^{desired} - \sum_k f_k u_{i-k})^2 \rightarrow \min$$

- Gives rise to a broad group of techniques:
 - e.g., for $u^{desired}$ being a spike, delayed spike, or a specified shape, this gives *spiking*, *optimal*, or *shaping* deconvolution

Deconvolution

Prediction-error (or "predictive")

- Time-domain
- Constructs a filter predicting the wavelet from its preceding values:

$$w_i = \sum_k f_k w_{i-k}$$

- Then, "prediction-error" filter:

$$f_k^{PE} = \delta_{k,0} - f_k$$

removes the reverberation from the signal.

- To find the predictive filter $f_{k'}$, note its action on the auto-correlation of the wavelet ϕ :

$$\phi_i = \sum_k f_k \phi_{i-k} \quad (*)$$

- Wavelet's auto-correlation is approximately equal the total signal auto-correlation (the "white reflectivity" hypothesis)
- From "normal equations" (*), f_k is obtained.

Deconvolution

F-X (predictive in space domain)

- X- or XY-domain
- Operates for each frequency independently
- Note that any linear event...

$$u(x, t) = \delta(a + bx - t)$$

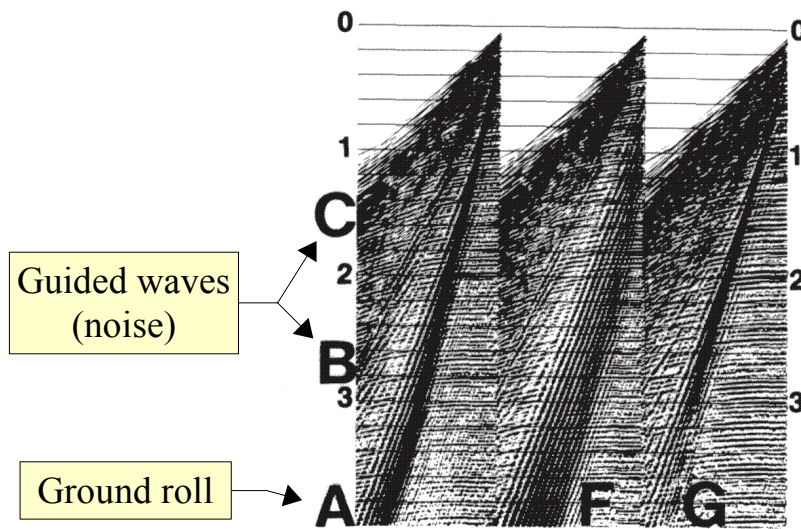
- ...after Fourier transform, becomes periodic in X:

$$u(x, \omega) = e^{i\omega a} e^{i\omega b x}$$

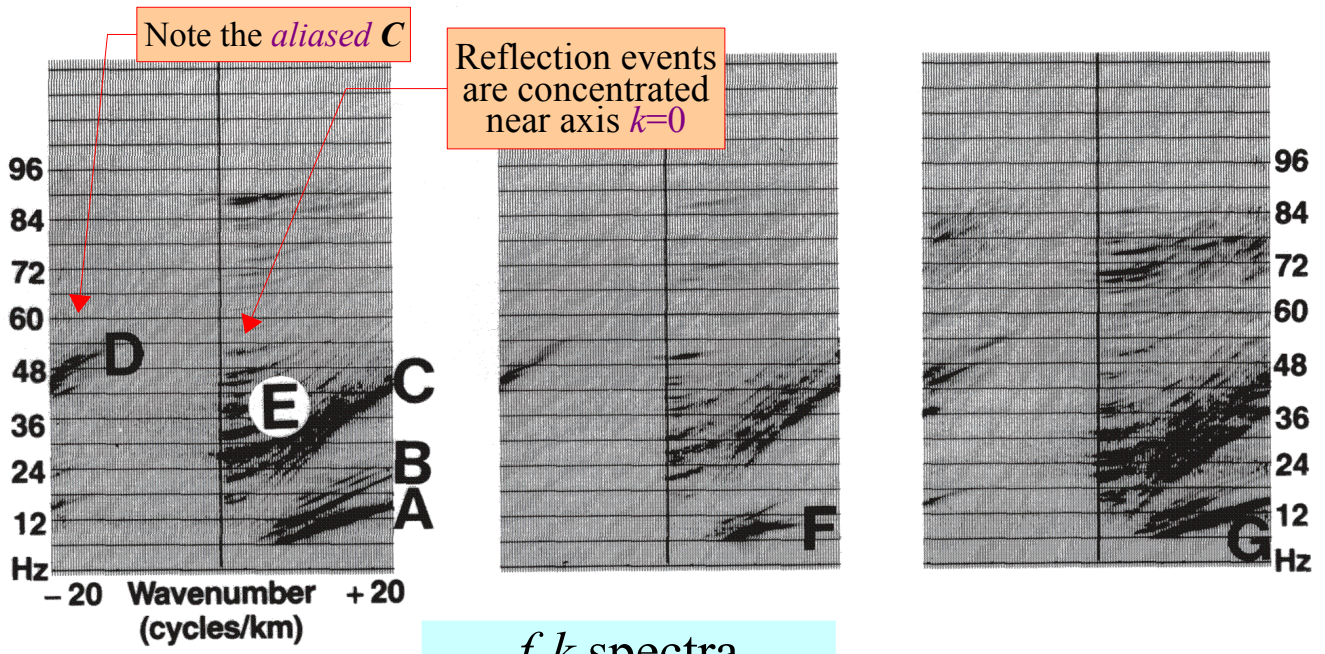
- Such **periodic events can be enhanced** by a predictive filter in X.
- Application:
 - Partition the data into windows small enough for the events to appear linear;
 - Fourier transform each window;
 - Calculate two prediction filters: one forward and one backward in X;
 - Sum the two predictions and transform back into the time domain.

F-K spectra (shot gathers)

- By performing Fourier Transform in both time and space, the *f-k spectra* are obtained
- The physical significance is in decomposition of the wavefield into *harmonic plane waves*



Shot gathers

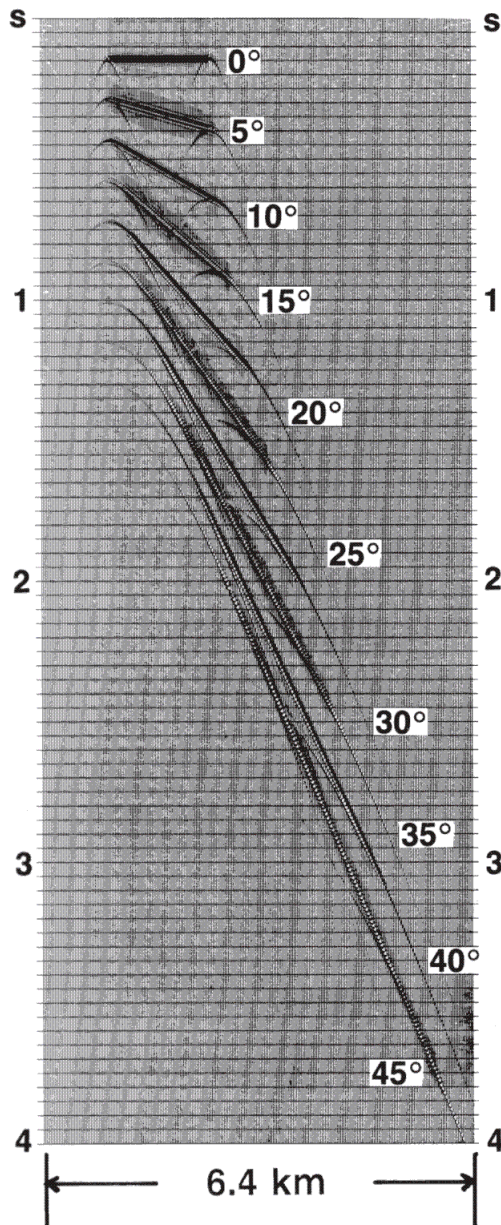


f-k spectra of the same gathers

From Yilmaz, 1987

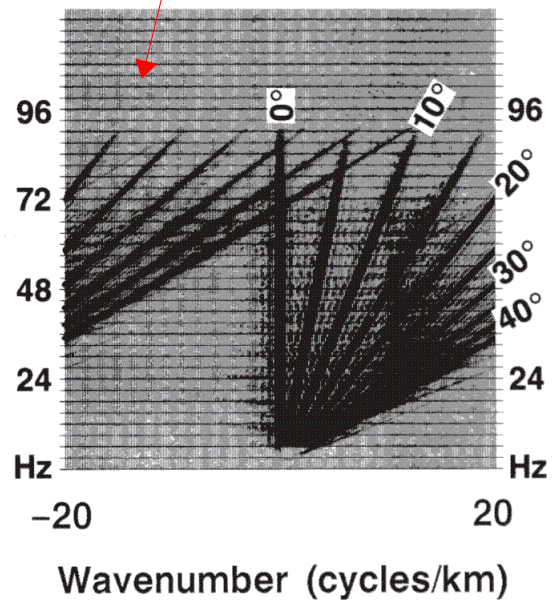
F-K spectra (dipping events in a zero-offset section)

- Events with different (apparent) dips occupy different parts of the $f-k$ spectrum, regardless of their positions in time or space



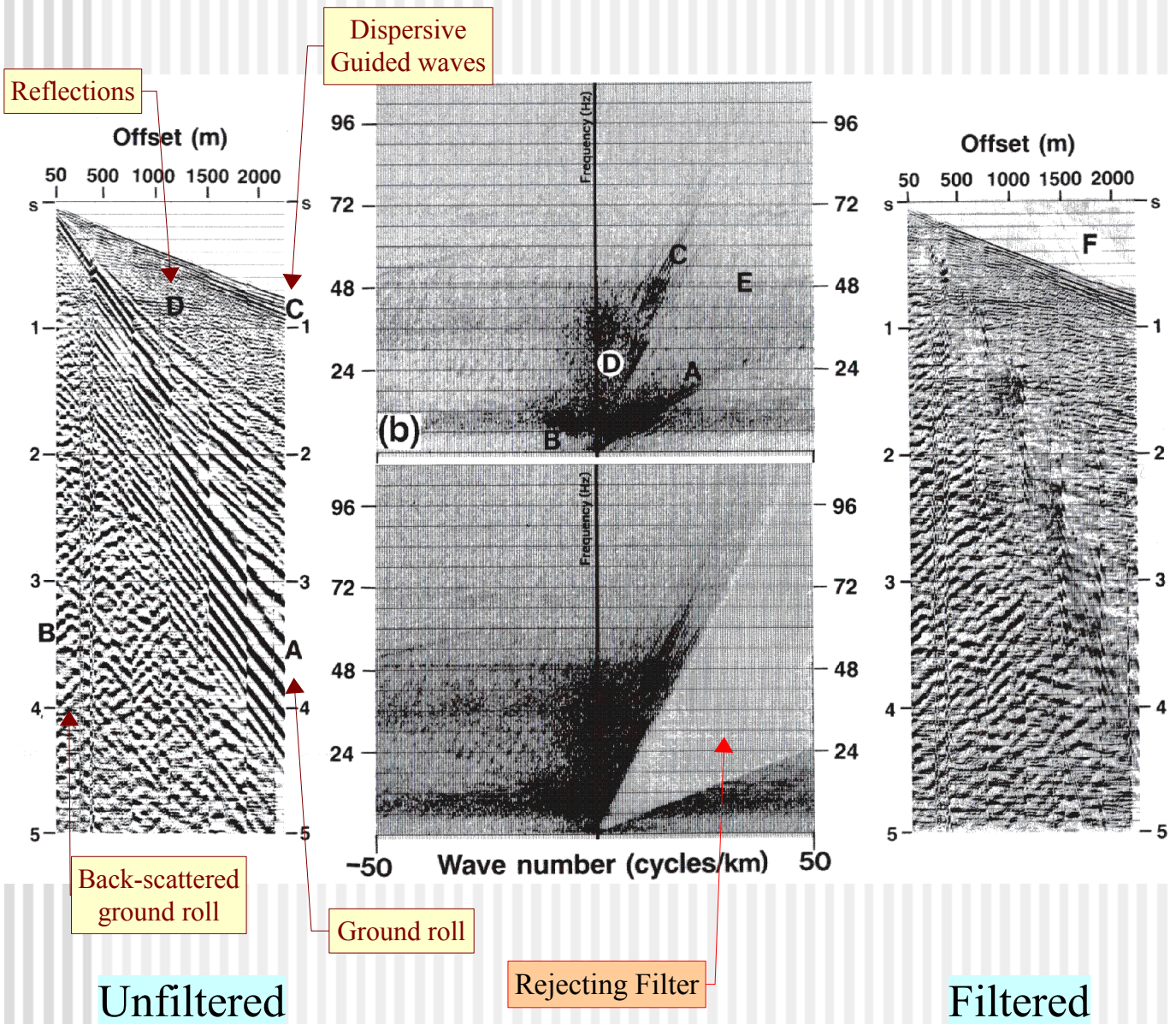
From Yilmaz, 1987

With increasing dips, the events become *aliased* at progressively lower frequencies



F-K filtering

- Here, only forward-propagating ground roll is rejected by the filter.



Plane-wave decomposition

t-p transform

- Instead of *f-k* transform, plane waves can be extracted from the section by *slant-stacking*:

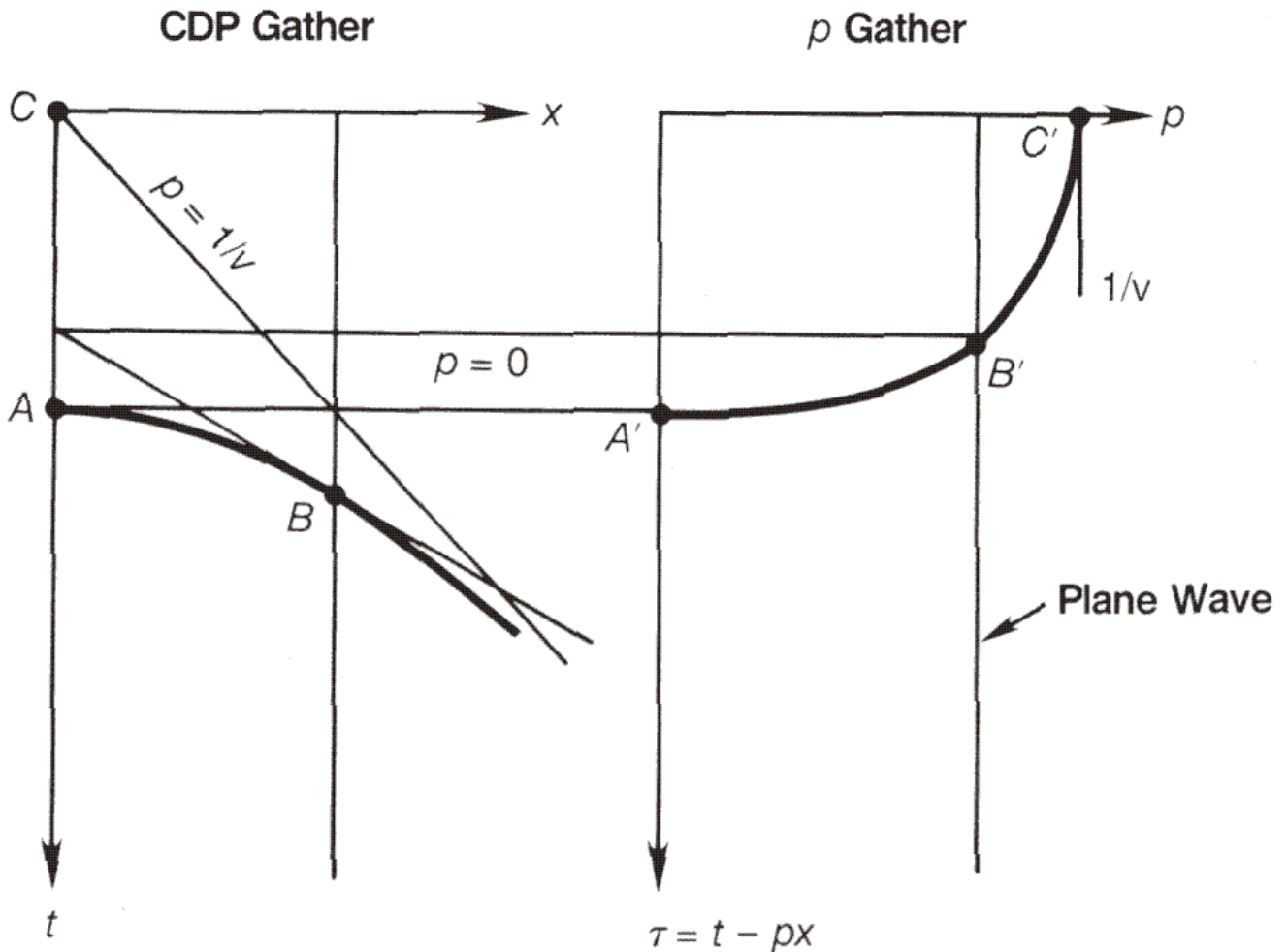
$$S(p, \tau) = \sum_x u(x, \tau + px)$$

$t = \tau + px$
describes the wavefront
of a plane wave

- This is done for every τ (intercept time) and p (slowness), resulting in a (τ, p) section
- The difference from *f-k* is in using plane waves *localized in time* (pulses instead of harmonic functions),
 - ...and therefore filtering can be based on *moveouts AND times* of the events.

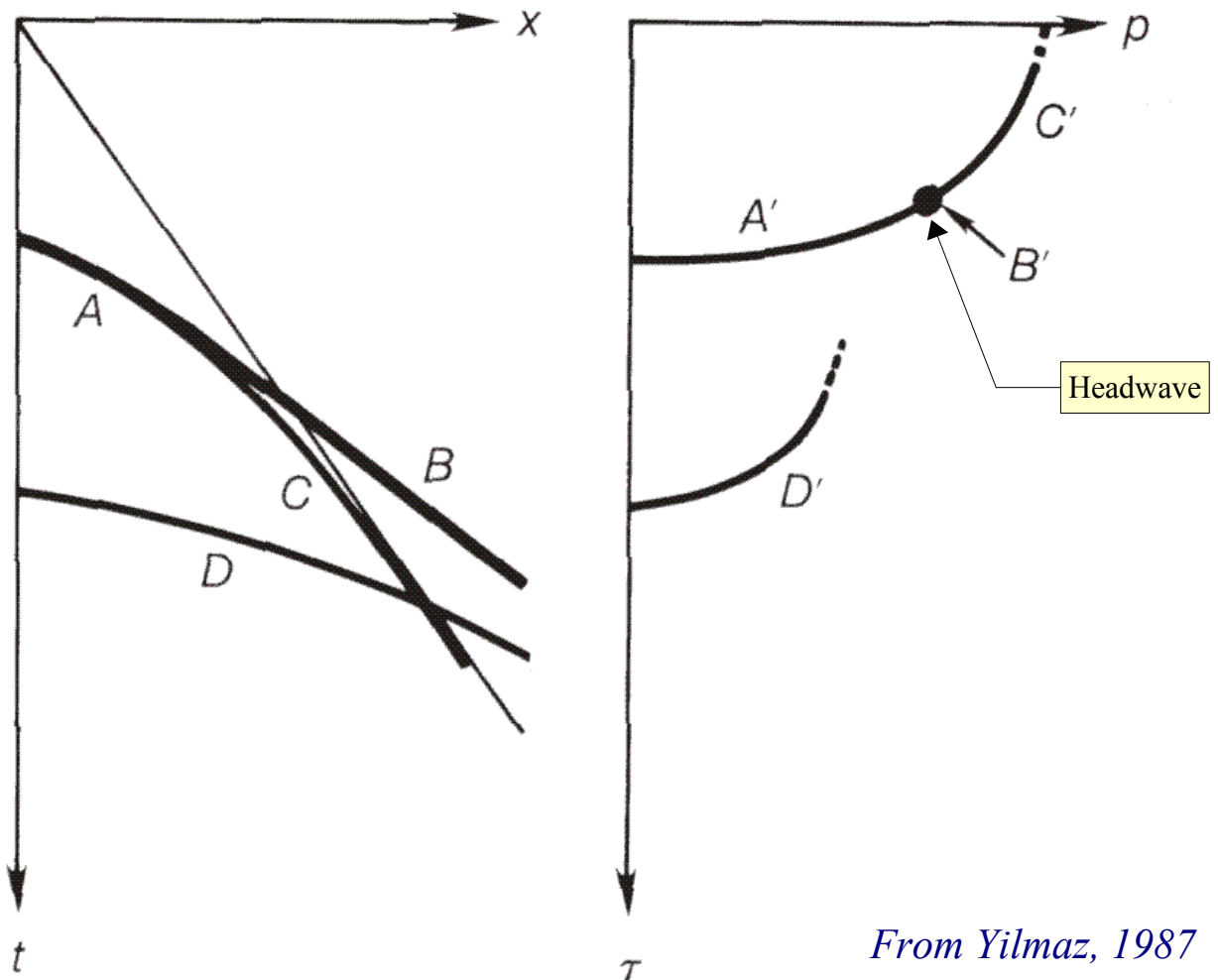
Refractions and reflections in τ - p domain

- Reflections (straight lines in (x,t)) become points,
- ...and reflections (hyperbolas in (x,t)) - ellipses



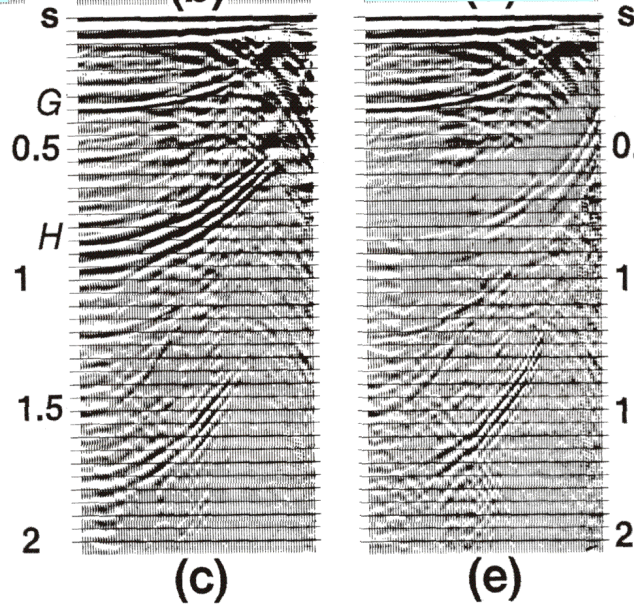
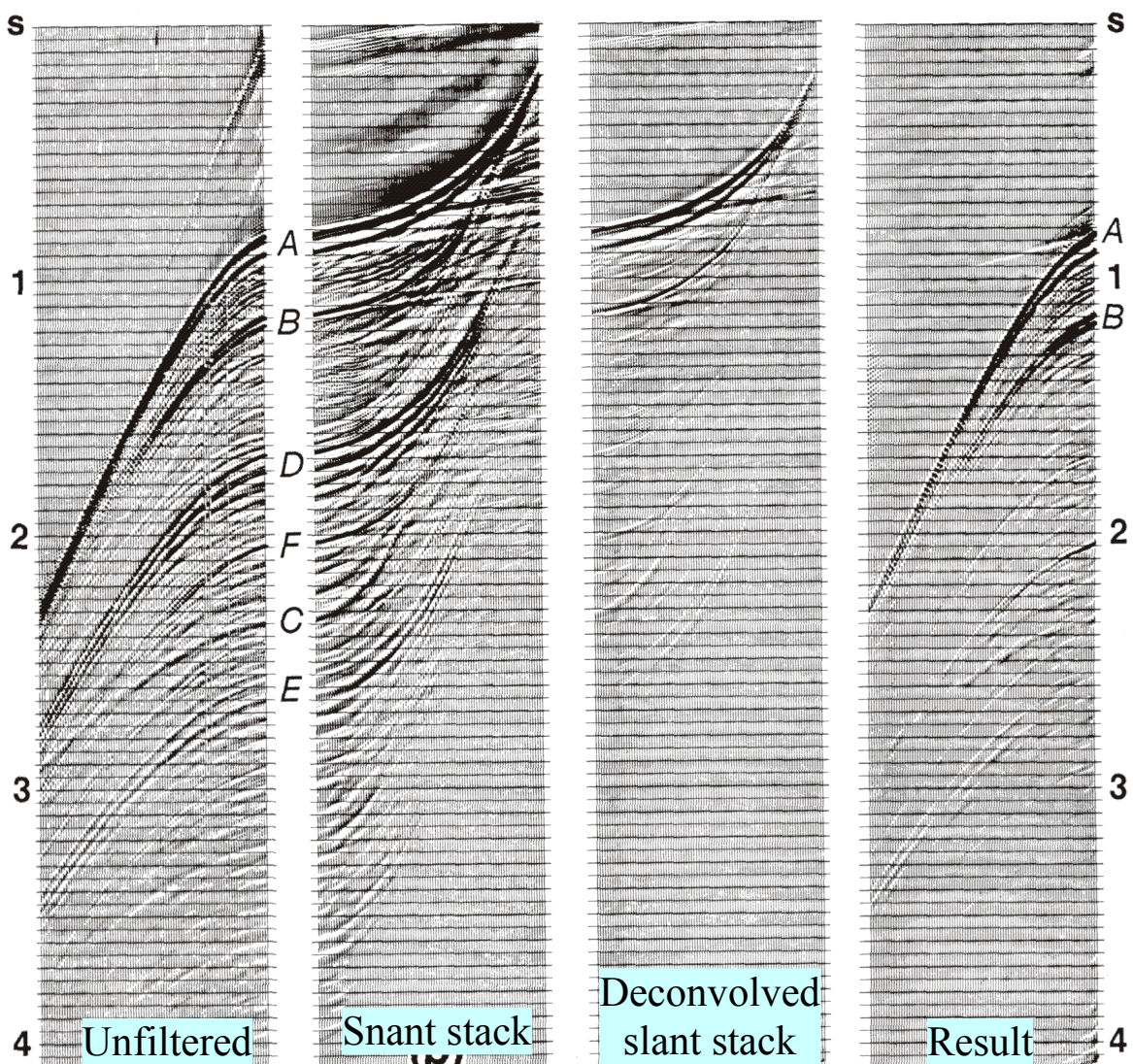
Several reflections in τ - p domain

- Reflections can be separated by their intercept times
- Phases retain their waveforms – this simplifies interpretation and facilitates waveform shaping (e.g., deconvolution)



From Yilmaz, 1987

Multiple suppression using τ - p



These are autocorrelations of the τ -traces above. Note how deconvolution removes the reverberations

From Yilmaz, 1987