

the maximum frequency in the signal. Thus provided the Fourier transform of the signal is nonzero only for  $f \in (-1/2, 1/2)$ , the signal can be reconstructed from its samples using the sampling expansion or by low pass filtering the a signal with the samples imbedded on impulses.

- A difference equation defines a linear system, and the system is time invariant since the coefficients are. The system can be characterized as convolving the input with the response to a Kronecker delta  $x[n] = \delta(n) = 1$  for  $n = 0$ , and 0 otherwise. Since the system is assumed causal,  $y[n] = 0$  for  $n < 0$  and hence  $y[0] = 1$ ,  $y[1] = a$ ,  $y[2] = a^2$ , etc. so that the response to a Kronecker delta is  $h[n] = a^n$  for  $n = 0, 1, 2, \dots$  and 0 otherwise. The inverse filter has Kronecker delta response  $g[n] = \delta[n] - a\delta[n - 1]$ , the discrete time convolution of  $h$  and  $g$  is  $\delta$ . Alternatively, from the geometric series (assuming  $|a| < 1$ )

$$H(f) = \sum_{n=0}^{\infty} a^n e^{-j2\pi f n} = \frac{1}{1 - ae^{-j2\pi f}}$$

and the inverse filter is  $1/H(f)$ .

Alternatively, taking the transform of the difference equation and changing variables (or using the delay theorem)

$$\begin{aligned} Y(f) &= \sum_{n=-\infty}^{\infty} y[n] e^{-j2\pi f n} \\ &= \sum_{n=-\infty}^{\infty} ay[n - 1] e^{-j2\pi f n} + \sum_{n=-\infty}^{\infty} x[n] e^{-j2\pi f n} \\ &= \sum_{n'=-\infty}^{\infty} ay[n'] e^{-j2\pi f (n'+1)} + X(f) \\ &= e^{-j2\pi f} Y(f) + X(f) \end{aligned}$$

so that

$$Y(f) = \frac{X(f)}{1 - e^{-j2\pi f}}$$

If  $a = 1$ , then the first argument still works and the system is still invertible, but the transform arguments get more complicated.