

Solutions

Problem 1

Multiplying the phase of the Fourier transform by -1 amounts to conjugating the spectrum; hence,

$$G(s) = F^*(s).$$

Therefore

$$g(t) = f^*(-t).$$

If the input is real-valued, the output is a time-reversed version. The system is non-causal, non-linear, and time variant. If the input is constrained to be real valued, the system behaves linearly.

Problem 2

- (a) Yes, one can throw away every other sample and still exceed the Nyquist frequency.
 - (b) Yes, model the "defective" sampling by $f(t)[1/T \text{comb}(t/T) - \delta(t)]$. The transform of this quantity is the original replicated spectrum shifted vertically based on the value $f(0)$. Therefore, $f(0)$ can be determined by evaluating the spectrum in the gaps between replication islands since the values there should be zero. A time domain approach is possible too.
 - (c) No, not possible because of aliasing.
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