EE 508 Lecture 12

The Approximation Problem

Classical Approximating Functions - Thomson and Bessel Approximations

Statistical Characterization of Filter Circuits

Review from Last Time

Thomson and Bessel Approximations

- All-pole filters
- Maximally linear phase at ω =0

Consider T(jω)

$$T(j\omega) = \frac{N(j\omega)}{D(j\omega)} = \frac{N_R(\omega) + jN_{IM}(\omega)}{D_R(\omega) + jD_{IM}(\omega)}$$
phase = $\angle (T(j\omega)) = \tan^{-1} \left(\frac{N_{IM}(\omega)}{N_R(\omega)}\right) - \tan^{-1} \left(\frac{D_{IM}(\omega)}{D_R(\omega)}\right)$

- Phase expressions are difficult to work with
- Will first consider group delay and frequency distortion

Review from Last Time

Linear Phase

Consider T(jω)

$$T(j\omega) = \frac{N(j\omega)}{D(j\omega)} = \frac{N_R(\omega) + jN_{IM}(\omega)}{D_R(\omega) + jD_{IM}(\omega)}$$
$$\angle (T(j\omega)) = \tan^{-1} \left(\frac{N_{IM}(\omega)}{N_R(\omega)}\right) - \tan^{-1} \left(\frac{D_{IM}(\omega)}{D_R(\omega)}\right)$$

Defn: A filter is said to have linear phase if the phase is given by the expression $\angle (T(j\omega)) = \Theta \omega$ where Θ is a constant that is independent of ω

Review from Last Time

Preserving wave-shape in pass band

A filter is said to have linear passband phase if the phase in the passband of the filter is given by the expression $\angle (T(j\omega)) = \Theta \omega$ where Θ is a constant that is independent of ω

If a filter has linear passband phase in a flat passband, then the waveshape is preserved provided all spectral components of the input are in the passband and the output can be expressed as an amplitude scaled and time shifted version of the input by the expression

$$V_{OUT}(t) = KV_{IN}(t-t_{shift})$$

Amplitude (Magnitude) Distortion, Phase Distortion and Preserving wave-shape in pass band

$$X_{IN}(s) \rightarrow T(s) \rightarrow T(s)$$

Amplitude and phase distortion are often of concern in filter applications requiring a flat passband and a flat zero-magnitude stop band

Amplitude distortion is usually of little concern in the stopband of a filter

Phase distortion is usually of little concern in the stopband of a filter

A filter with no amplitude distortion or phase distortion in the passband and a zero-magnitude stop band will exhibit waveform distortion for any input that has a frequency component in the passband and another frequency component in the stopband

It can be shown that the only way to avoid magnitude and phase distortion respectively for signals that have energy components in the interval $\omega_1 < \omega < \omega_2$ is to have constants k_1 and k_2 such that

$$\begin{array}{c} \mathsf{T}(\mathsf{j}\omega) \big| = \mathsf{k}_1 \\ \angle \mathsf{T}(\mathsf{j}\omega) = \mathsf{k}_2 \omega \end{array} \right\} \qquad \text{for } \omega_1 < \omega < \omega_2 \end{array}$$

Review from Last Time

Group Delay

Defn: Group Delay is the negative of the phase derivative with respect to $\boldsymbol{\omega}$

$$\tau_{G} = -\frac{\mathsf{d}\angle\mathsf{T}(\mathsf{j}\omega)}{\mathsf{d}\omega}$$

Recall, by definition, the phase is linear iff $\angle T(j\omega) = k\omega$

If the phase is linear,
$$\tau_G = -\frac{d\angle T(j\omega)}{d\omega} = -\frac{d(k\omega)}{d\omega} = -k$$

Thus for $\angle T(j0) = 0$, the phase is linear iff the group delay is constant

The group delay and the phase of a transfer function carry the same information

But, of what use is the group delay?

But, of what use is the group delay?

The phase of almost all useful transfer functions are complicated functions involving sums of arctan functions and these are difficult to work with analytically

Theorem: The group delay of any transfer function is a rational fraction in ω^2

But, of what use is the group delay?

Qualitatively:

The following two criteria are equivalent:

- Maximally linear phase at ω =0
- Maximally constant group delay at ω =0

Analytically working with the group delay (rational fraction in ω^2) rather than the phase (difference between 2 arctan functions) is much more mathematically tractable

Theorem: The group delay of any transfer function is a rational fraction in ω^2

Proof of Theorem (only shown here for case of all-pole transfer function):

$$T(s) = \frac{1}{\sum_{k=0}^{n} a_k s^k}$$

$$T(j\omega) = \frac{1}{(1 - a_2\omega^2 + a_4\omega^4 + ...) + j\omega(a_1 - a_3\omega^2 + a_5\omega^4 + ...)}$$

$$T(j\omega) = \frac{1}{F_1(\omega^2) + j\omega F_2(\omega^2)}$$

where F_1 and F_2 are even polynomials in ω

$$\angle T(j\omega) = -\tan^{-1}\left(\frac{\omega F_2(\omega^2)}{F_1(\omega^2)}\right)$$

Theorem: The group delay of any transfer function is a rational fraction in ω^2

Proof of Theorem:
$$\angle T(j\omega) = -\tan^{-1}\left(\frac{\omega F_2(\omega^2)}{F_1(\omega^2)}\right)$$

but from identity $\frac{d(\tan^{-1}u)}{dx} = \left(\frac{1}{1+u^2}\right)\frac{du}{dx}$
 $\tau_G = -\frac{d\angle T(j\omega)}{d\omega} = -\frac{1}{1+\left[\frac{\omega F_2(\omega^2)}{F_1(\omega^2)}\right]^2} \bullet \frac{d\left[\frac{\omega F_2(\omega^2)}{F_1(\omega^2)}\right]}{d\omega}$

Now consider the right-most term in the product

$$\frac{d\left[\frac{\omega F_{2}(\omega^{2})}{F_{1}(\omega^{2})}\right]}{d\omega} = \frac{F_{1}(\omega^{2})\left[\frac{d\left(\omega F_{2}(\omega^{2})\right)}{d\omega}\right] - \left(\omega F_{2}(\omega^{2})\right)\frac{d\left(F_{1}(\omega^{2})\right)}{d\omega}}{\left[F_{1}(\omega^{2})\right]^{2}}$$

Theorem: The group delay of any transfer function is a rational fraction in ω^2



Thus this term is an even rational fraction in ω

Theorem: The group delay of any transfer function is a rational fraction in ω^2



It follows that T_G is the product of rational fractions in ω^2 so it is also a rational fraction in ω^2

Although tedious, the results can be extended when there are zeros present in T(s) as well

- All-pole filters
- Maximally linear phase at $\omega=0$

$$- \frac{d\angle T(j\omega)}{d\omega}\bigg|_{\omega=0} = -1$$

 $\tau_G = -\frac{\mathbf{u} \mathbf{r} (\mathbf{j} \mathbf{w})}{\mathbf{d} \mathbf{w}}$

since

These criteria can be equivalently expressed as

- All-pole filters
- Maximally constant group delay at $\omega=0$
- $\tau_G = 1$ at $\omega = 0$

$$\mathsf{T}_{\mathsf{A}}(\mathsf{s}) = \frac{1}{\sum_{k=0}^{n} \mathsf{a}_{k} \mathsf{s}^{k}}$$

Must find the coefficients $a_0, a_1, \dots a_n$ to satisfy the maximal constant group delay constraints

$$T(j\omega) = \frac{1}{(1 - a_2\omega^2 + a_4\omega^4 + ...) + j\omega(a_1 - a_3\omega^2 + a_5\omega^4 + ...)}$$

Theorem: If $T(j\omega) = \frac{1}{x + jy}$ then τ_G is given by the expression $\tau_G = \frac{x \frac{dy}{d\omega} - y \frac{dx}{d\omega}}{x^2 + y^2}$

This theorem is easy to prove using the identity given above, proof will not be given here

$$\mathsf{T}_{\mathsf{A}}(\mathsf{s}) = \frac{1}{\sum_{k=0}^{n} \mathsf{a}_{k} \mathsf{s}^{k}}$$

Must find the coefficients $a_0, a_1,...$ an to satisfy the constraints

$$T(j\omega) = \frac{1}{(1 - a_2\omega^2 + a_4\omega^4 + ...) + j\omega(a_1 - a_3\omega^2 + a_5\omega^4 + ...)}$$

From this theorem, it follows that

$$\tau_{G} = \frac{a_{1} + \omega^{2} (a_{1}a_{2} - 3a_{3}) + \omega^{4} (5a_{5} - 3a_{1}a_{4} + a_{2}a_{3}) + \dots}{1 + \omega^{2} (a_{1}^{2} - 2a_{2}) + \omega^{4} (a_{2}^{2} - 2a_{1}a_{3} + 2a_{4}) + \dots}$$

from the constraint $\tau_G = 1$ at ω =0, it follows that a_1 =1

To make τ_G maximally constant at ω =0, want to match as many coefficients in the numerator and denominator as possible starting with the lowest powers of ω^2 from ω^2 terms $a_1a_2-3a_3 = a_1^2 - 2a_2$ from ω^4 terms $5a_5-3a_1a_4+a_2a_3 = a_2^2-2a_1a_3+2a_4$

$$\mathsf{T}_{\mathsf{A}}(\mathsf{s}) = \frac{1}{\sum_{k=0}^{n} \mathsf{a}_{k} \mathsf{s}^{k}}$$

Must find the coefficients a_0, a_1, \dots, a_n to satisfy the constraints

It can be shown that the a_k 's are given by

$$a_{k} = \frac{(2n-k)!}{H2^{n-k}k!(n-k)!}$$
$$a_{n} = H$$

for $1 \le k \le n-1$

where

$$H = \frac{(2n)!}{2^n n!}$$

Note that all coefficients are real !

Inverse mapping thus exists !

$$\mathsf{T}_{\mathsf{A}}(\mathsf{s}) = \frac{1}{\sum_{k=0}^{n} \mathsf{a}_{k} \mathsf{s}^{k}}$$

Must find the coefficients $a_0, a_1,...$ an to satisfy the constraints

Alternatively, if we define the recursive polynomial set by

$$B_{1} = s+1$$

$$B_{2} = s^{2} + 3s + 3$$

...

$$B_{k} = (2k-1)B_{k-1} + s^{2}B_{k-2}$$

Then the n-th order Thompson approximation is given by

$$T_{An}(s) = \frac{B_{n}(0)}{B_{n}(s)}$$

Since the recursive set of polynomials are termed Bessel functions, this is often termed the Bessel approximation

The poles of the BW and CC approximations were obtained analytically

What are the poles of the Thomson approximation?

The Thomson approximation directly results in a polynomial in s rather than a set of poles

If is straightforward to analytically obtain a rational fraction if the poles and zeros are known but analytically obtaining the poles and zeros from an arbitrary rational fraction is not possible for 5th and higher order systems



Friedrich Bessel 1784-1846 Astronomer, Physicist, Mathematician

Was Bessel before his time in the filter field?

$$T_{An}(s) = \frac{B_{n}(0)}{B_{n}(s)}$$



http://www.rfcafe.com/references/electrical/bessel-poles.htm

- Poles of Bessel Filters lie on circle
- Circle does not go through the origin
- Poles not uniformly space on circumference

$$T_{An}(s) = \frac{B_{n}(0)}{B_{n}(s)}$$

Observations:

The Thomson approximation has relatively poor magnitude characteristic (at least if considered as an approximation to the standard lowpass function)

The normalized Thomson approximation has a group delay of 1 or a phase of ω at $\omega\text{=}0$

Frequency scaling is used to denormalize the group delay or the phase to other values

Use of Bessel Filters:

$$X_{IN}(s) \rightarrow T(s) \xrightarrow{X_{OUT}(s)}$$

Consider: $T(s) = e^{-sh}$ (not realizable but can be approximated) $T(j\omega) = e^{-j\omega h}$ $T(j\omega) = \cos(-\omega h) + j\sin(-\omega h)$ $|T(j\omega)| = 1$ $\angle T(j\omega) = -h\omega$ If $x_{IN}(t) = X_M \sin(\omega t + \theta)$ $x_{OUT}(t) = X_{M} sin(\omega t + \theta - h\omega)$ $x_{OUT}(t) = X_{M} sin(\omega[t-h]+\theta)$ This is simply a delayed version of the input

$$\boldsymbol{x}_{\text{OUT}}(t) = \boldsymbol{x}_{\text{IN}}(t-h)$$

But
$$\boldsymbol{\tau}_{G} = \frac{-d \angle \mathsf{T}(j\omega)}{d\omega} = h \qquad \boldsymbol{x}_{\text{OUT}}(t) = \boldsymbol{x}_{\text{IN}}(t-\tau_{G})$$

So, output is delayed version of input and the delay is the group delay



It is challenging to build filters with a constant delay

A filter with a constant group delay and unity magnitude introduces a constant delay

Bessel filters are filters that are used to approximate a constant delay

Bessel filters are attractive for introducing constant delays in digital systems

Some authors refer to Bessel filters as "Delay Filters"

An ideal delay filter would

- introduce a time-domain shift of a step input by the group delay
- introduce a time-domain shift of each spectral component by the group delay
- introduce a time-domain shift of a square wave by the group delay



Characterization of the step response of a filter

From Introduction to the Theory and Design of Active Filters by Huelsman and Allen, p. 94-96



Step Response of Butterworth Filter

Delay is not constant Overshoot present and increases with order BW filters do not perform well as delay filters

From Introduction to the Theory and Design of Active Filters by Huelsman and Allen, p. 94-96



Step Response of Chebyschev Filter

Delay is not constant

Overshoot and ringing present and increases with order CC filters do not perform well as delay filters



Step Response of Bessel Filters

Delay becomes more constant as order increases No overshoot or ringing present Bessel filters widely used as delay filters Bessel filters often designed to achieve time-domain performance

From Introduction to the Theory and Design of Active Filters by Huelsman and Allen, p. 94-96



- Comparison of Step Response of 3rd-order Bessel, BW and CC filters
- Comparison for normalized frequency response for BW, CC and normalized group delay for Bessel



Figure 10.3 Delay of Bessel-Thomson filters of orders 2 through 10.



Harmonics in passband of Bessel Filter increase with n

Attenuation of amplitude for Bessel does not compare favorably wth BW, CC, or Eliptic filters

Figure 10.4 Comparison of Bessel–Thomson and Butterworth responses of orders 4 and 8.



Magnitude of Bessel filters does not drop rapidly at band edge

Phase of Bessel filters becomes very linear in passband as order increases

From Continuous-Time Active Filter Design by Deliyannis, Sun and Fidler



Comparison of Phase Response of 3rd-order Bessel, BW and CC filters

Statistical Characterization of Filter Characteristics

Components used to build filters are not precisely predictable



- Temperature Variations
- Manufacturing Variations
- Aging
- Model variations

Different approaches are used to address each of these problems
 Manufacturing variations is one of the most challenging problems for building integrate filters and will be the focus of this lecture

Wafers are processed in "batches" or "lots" of 20 to 40 wafers and variations occur over time (process not completely stationary) and over location









 $R(t_3)$

These variations are often the major contributor to process variability and can be in the $\pm 30\%$ range or larger

These variations often look like random variations



Stay Safe and Stay Healthy !

End of Lecture 12