# Poles/Zeros/Gain from Look-Up Tables (Prototyping)

BSM Oct 2020

```
In [1]: import numpy as np
    from scipy import signal
    import vectfit
    import matplotlib.pyplot as plt
    from matplotlib.gridspec import GridSpec
%matplotlib inline
```

```
In [2]: # NIMS FILTERS
        # These values are from here:
        # http://service.iris.edu/fdsnws/station/1/query?net=EM&sta=IDA11&cha=
        *&level=response&format=xml&nodata=404
        # ALSO can pull these from an sp file (qains are clearer in sp files)
        # B fields, one filter:
        zeros_Blowpass = []
        poles_Blowpass = [-6.28319 + 10.8825j, -6.28319 - 10.8825j, -12.5664]
        gain Blowpass = 1984.31
        NIMS Magnetic 3PoleLowpass = \
            signal.ZerosPolesGain(zeros Blowpass, poles Blowpass, gain Blowpass)
        # E fields, two filters:
        zeros Ehighpass = [0.]
        poles Ehighpass = [-1.66667E-4]
        gain Ehighpass = 1.
        NIMS Electric Highpass = \
            signal.ZerosPolesGain(zeros Ehighpass, poles Ehighpass,
                                  gain Ehighpass)
        zeros Elowpass = []
        poles_Elowpass = [-3.88301 + 11.9519j, -3.88301 - 11.9519j,
                          -10.1662 + 7.38651j, -10.1662 - 7.38651j,
                          -12.5664
        gain Elowpass = 313384.
        NIMS Electric 5PoleLowpass = \
            signal.ZerosPolesGain(zeros Elowpass, poles Elowpass, gain Elowpass)
```

```
In [3]: # ZEN COIL RESPONSE
        # angular frequency, real, imaginary
        zen_response = np.array(
             [[0.0006283185307179586,\ 0.00016610681887368792,\ 0.2785499504730251] 
        ],
             [0.0012566370614359172, 0.009607482581442478, 0.5968226757408328],
             [0.0031415926535897933, 0.0340338255267764, 1.1534980271851412],
             [0.006283185307179587, 0.10326125512223676, 2.345528075549423],
             [0.012566370614359173, 0.2694037321803452, 4.648009077571525],
             [0.024504422698000385, 0.8674625452867567, 9.250988152485366],
             [0.04900884539600077, 3.3996002487875288, 18.026842965656627],
             [0.09801769079200154, 12.288173305808309, 32.68634912660055],
             [0.19666370011472106, 35.77374451709642, 47.65907329465832],
             [0.39269908169872414, 68.80622670880321, 45.83081507358207],
             [0.7853981633974483, 89.45957701017464, 29.675744159172037],
             [1.5707963267948966, 96.71915794659112, 15.82688714795047],
             [3.141592653589793, 98.73350517756762, 7.627472850645089],
             [6.283185307179586, 99.24219816830174, 2.958293251509568],
             [12.566370614359172, 99.30832240704096, -0.2879949423266931],
             [25.132741228718345, 99.24383148030137, -3.7035126269049026],
             [50.26548245743669, 98.97856371276684, -8.952157997609012],
             [100.53096491487338, 97.62982613999657, -18.64457201832332],
             [201.06192982974676, 92.02668043950737, -37.226970582652726],
             [402.1238596594935, 69.87783035350839, -70.17219305236156],
             [804.247719318987, -13.629323102806389, -89.21119896277264],
             [1608.495438637974, -26.895281950721987, 4.2573962015260145],
             [3216.990877275948, -2.138888186949132, 2.88837534502174],
             [6433.981754551896, -0.09177331053666367, 0.4396128837660938]]
        zen angular frequencies = zen response[:, 0]
        zen period = 2. * np.pi / zen angular frequencies
        zen complex response = zen response[:, 1] + 1.j * zen response[:, 2]
        zen amplitude response = np.absolute(zen complex response)
        zen phase response = np.angle(zen complex response, deg=True)
```

```
In [4]: def plot_response(w_obs=None, resp_obs=None, zpk_obs=None,
                           zpk pred=None, w values=None, xlim=None):
            fig = plt.figure(figsize=(14, 4))
            gs = GridSpec(2, 3)
            ax_amp = fig.add_subplot(gs[0, :2])
            ax phs = fig.add subplot(gs[1, :2])
            ax_pz = fig.add_subplot(gs[:, 2], aspect='equal')
            if w obs is not None and resp obs is not None:
                ax amp.plot(2. * np.pi / w obs, np.absolute(resp obs),
                            color='tab:blue', linewidth=1.5, linestyle='-',
                            label='True')
                ax phs.plot(2. * np.pi / w_obs, np.angle(resp_obs, deg=True),
                            color='tab:blue', linewidth=1.5, linestyle='-')
            elif zpk_obs is not None:
                w_obs, resp_obs = signal.freqresp(zpk_obs, w=w_values)
                ax amp.plot(2. * np.pi / w obs, np.absolute(resp obs),
                            color='tab:blue', linewidth=1.5, linestyle='-',
                            label='True')
                ax phs.plot(2. * np.pi / w obs, np.angle(resp obs, deg=True),
                            color='tab:blue', linewidth=1.5, linestyle='-')
                ax pz.scatter(np.real(zpk obs.zeros), np.imag(zpk obs.zeros),
                               s=75, marker='o', ec='tab:blue', fc='w',
                              label='True Zeros')
                ax pz.scatter(np.real(zpk obs.poles), np.imag(zpk obs.poles),
                              s=75, marker='x', ec='tab:blue', fc='tab:blue',
                              label='True Poles')
            if zpk_pred is not None:
                w pred, resp pred = signal.freqresp(zpk pred, w=w values)
                ax_amp.plot(2. * np.pi / w_pred, np.absolute(resp_pred),
                            color='tab:red',linewidth=3, linestyle=':',
                            label='Fit')
                ax_phs.plot(2. * np.pi / w_pred, np.angle(resp_pred, deg=True),
                            color='tab:red', linewidth=3, linestyle=':')
                ax pz.scatter(np.real(zpk pred.zeros), np.imag(zpk pred.zeros),
                               s=35, marker='o', ec='tab:red', fc='w',
                              label='Fit Zeros')
                ax_pz.scatter(np.real(zpk_pred.poles), np.imag(zpk_pred.poles),
                            s=35, marker='x', ec='tab:red', fc='tab:blue',
                           label='Fit Poles')
            if xlim is not None:
                ax amp.set xlim(xlim)
                ax phs.set xlim(xlim)
            ax amp.set xscale('log')
            ax amp.set yscale('log')
            ax amp.set ylabel('Amplitude Response')
            ax amp.grid()
            ax amp.legend()
            ax_phs.set_ylim([-180., 180.])
            ax phs.set xscale('log')
            ax phs.set ylabel('Phase Response')
```

# **Non-Linear Least Squares Fitting**

Using nonlinear least squares with Levenberg-Marquardt step length damping to fit poles and zeros representation to look-up table of complex system response. Nominally, the target function is of the form

$$zpk(s) = k \frac{(s-z_1)(s-z_2)(s-z_3)\dots(s-z_m)}{(s-p_1)(s-p_2)(s-p_3)\dots(s-p_n)}$$

where s is complex frequency ( $s = i\omega$ , where  $\omega$  is angular frequency), k is the overall system gain, z represents the m system zeros, and p represents the n system poles.

HOWEVER. We're actually going to fit the look-up table with the transfer function form of the system response function:

$$h(s) = \frac{a_m s^m + a_{m-1} s^{m-1} + \dots + a_1 s^1 + a_0}{b_n s^n + b_{n-1} s^{n-1} + \dots + b_1 s^1 + b_0}$$

where  $a_x$  and  $b_x$  are the (real) coefficients on the numerator and denominator polynomials, m is the number of system zeros (also the degree of the numerator polynomial), n is the number of system poles (also the degree of the denominator polynomial), and s is complex frequency ( $s = i\omega$ , where  $\omega$  in angular frequency).

This form has several benefits. First of all, because the coefficients are real, the roots of these polynomials (and therefore the poles/zeros) will occur as complex conjugate pairs, which is what we expect/require for a real linear system operating on a real time series (I think...). Having real coefficients in this polynomial form also makes the math a bit easier.

Then, to find the poles/zeros, we just have to factor the numerator and denominator of this rational function. Operationally, we'll just use Python tools to do so.

For the nonlinear least squares fitting, the Jacobian is fairly simple to calculate:

$$\frac{\partial h(s)}{\partial a_m} = \frac{s^m}{b_n s^n + b_{n-1} s^{n-1} + \dots + b_1 s^1 + b_0} = \frac{s^m}{d(s)}$$

(Other partial derivatives with respect to the numerator terms are similar.)

$$\frac{\partial h(s)}{\partial b_n} = -\frac{a_m s^m + a_{m-1} s^{m-1} + \dots + a_1 s^1 + a_0}{(b_n s^n + b_{n-1} s^{n-1} + \dots + b_1 s^1 + b_0)^2} s^n = -\frac{n(s)}{d(s)^2} s^n$$

(Other partial derivatives with respect to the denominator terms are similar.)

Here, n(s) and d(s) are the numerator and denominator polynomials, respectively.

In the matrices here, all the numerator terms will be listed first, then all the denominator terms.

```
In [7]: | def Fit ZerosPolesGain_toFrequencyResponse_LM(w, resp, m, n):
            w (array) : (real) angular frequencies at which we
                              have the system response
            resp (array) : complex system response at angular frequencies w
            m (int)
                        : number of zeros to use in the fit; number of
                              numerator coefficients will be m + 1
            n (int)
                        : number of poles to use in the fit; number of
                              denominator coefficients will be n + 1
            # set up initial quess.
            tf = np.zeros(m + 1 + n + 1, dtype=np.complex128)
            tf[:] = 1. # for now, just set initial guess at all ones
            print("Initial Numerator Guess: ", tf[:m+1])
            print("Initial Denominator Guess: ", tf[m+1:])
            # evaluate initial misfit
            resp pred = \
                evaluate transfer function response(tf[:m+1], tf[m+1:], w)
            resid = resp - resp pred
            misfit = np.mean(np.absolute(resid))
            print("Initial misfit: ", misfit)
            max iter = 10
            1 = 1
            for i in range(max iter):
                misfit previous = misfit
                # (1) get Jacobian matrix
                jac = get jacobian(tf[:m+1], tf[m+1:], w)
                # (2) solve the least squares system, with L-M damping
                hes = np.matmul(jac.conj().T, jac)
                hes += 1 * np.eye(m + 1 + n + 1)
                d tf = np.matmul(np.matmul(np.linalg.inv(hes), jac.conj().T),
                                 resid)
                # PROBLEM the polynomial coefficients are becoming complex
                # (3) update model parameter
                tf += d tf
                # (4) evaluate misfit
                resp pred = \
                    evaluate transfer function response(tf[:m+1], tf[m+1:], w)
                resid = resp - resp_pred
                misfit = np.mean(np.absolute(resid))
                print("Misfit Step {:d}: ".format(i+1), misfit)
                # (5) adjust damping term 1
                if misfit > misfit previous: 1 *= 2
                else: 1 /= 2
            return signal.TransferFunction(tf[:m+1], tf[m+1:]).to zpk()
```

#### Nonlinear LM fit to NIMS B response

-100

 $10^{-1}$ 

```
print("PARAMETERIZATION: 0 zeros, 3 poles")
In [8]:
         test period = np.logspace(-1, 5, num=100)
         test w = 2. * np.pi / test period
         NIMS w, NIMS Bresp = signal.freqresp(NIMS Magnetic 3PoleLowpass,
                                                 w=test w)
                                                               # w is in rad/sec
         NIMS LM ZPK = Fit ZerosPolesGain toFrequencyResponse LM(NIMS w,
                                                                       NIMS_Bresp,
                                                                       0, 3)
         plot response(zpk obs=NIMS Magnetic 3PoleLowpass, zpk pred=NIMS LM ZPK,
                        w_values=test_w)
         PARAMETERIZATION: 0 zeros, 3 poles
         Initial Numerator Guess: [1.+0.j]
         Initial Denominator Guess: [1.+0.j 1.+0.j 1.+0.j]
         Initial misfit:
                           0.3905301610061775
         Misfit Step 1:
                          0.3403990610683085
         Misfit Step 2:
                          0.364175109332949
         Misfit Step 3:
                          0.3060996107464751
         Misfit Step 4:
                          0.355562825616186
         Misfit Step 5:
                          0.28887850705540613
         Misfit Step 6:
                          0.24457928512349647
         Misfit Step 7:
                          0.21663084876634714
         Misfit Step 8:
                          0.1827350637992361
         Misfit Step 9:
                          0.1545248385960636
         Misfit Step 10:
                           0.15382987812226132
         Amplitude Response
                                                                 15
                                                                                 True Zeros
                                                           True
                                                        • • • Fit
                                                                               ×
                                                                                 True Poles
           10-2
                                                                 10
                                                                                 Fit Zeros
                                                                               0
                                                                                 Fit Poles
           10^{-4}
                                                              lm(z)
                                                           105
         hase Response
           100
                                                                 -5
                                                                -10
```

Ok, so the non-linear least squares approach isn't working so well. Apparently there's no guarantee that the model updates will be purely real (as I had hoped), and it's not converging very well. This could be because I used a bad initial guess, or because I'm doing something wrong. But let's try something else...

Period (s)

-15 -15 -10 -5

10

Re(z)

# **Linear Least Squares**

We can do some equation manipulation to turn this into a linear least squares problem. Again starting with the transfer function representation of the response function:

$$h(s) = \frac{a_m s^m + a_{m-1} s^{m-1} + \dots + a_1 s^1 + a_0}{b_n s^n + b_{n-1} s^{n-1} + \dots + b_1 s^1 + b_0} = \frac{n(s)}{d(s)}$$

where  $a_x$  and  $b_x$  are the (real) coefficients on the numerator and denominator polynomials, m is the number of system zeros (also the degree of the numerator polynomial), n is the number of system poles (also the degree of the denominator polynomial), and s is complex frequency ( $s = i\omega$ , where  $\omega$  in angular frequency).

Rearranging this equation...

$$d(s) \cdot h(s) = n(s)$$

$$(b_n s^n + b_{n-1} s^{n-1} + \dots + b_1 s^1 + b_0) \cdot h(s) = a_m s^m + a_{m-1} s^{m-1} + \dots + a_1 s^1 + a_0$$

If we require  $b_0=1$ , then we can get this into a convenient linear least squares form. (Note that  $b_0=1$  just means that all the other polynomial coefficients are effectively normalized by whatever  $b_0$  would've been before.)

$$b_n s^n \cdot h(s) + b_{n-1} s^{n-1} \cdot h(s) + \dots + b_1 s^1 \cdot h(s) + 1 \cdot h(s) = a_m s^m + a_{m-1} s^{m-1} + \dots + a_1 s^1 + a_0$$
$$h(s) = a_m s^m + a_{m-1} s^{m-1} + \dots + a_1 s^1 + a_0 - b_n s^n \cdot h(s) - b_{n-1} s^{n-1} \cdot h(s) - \dots - b_1 s^1 \cdot h(s)$$

So now we can have the complex response values h(s) in the RHS vector  $\mathbf{d}$ ; the coefficients  $a_m$  and  $b_m$  as the model parameter  $\mathbf{m}$ ;  $s^m$  and  $-s^n \cdot h(s)$  in the design matrix  $\mathbf{G}$ .

d = Gm

$$\begin{bmatrix} h(s_0) \\ h(s_1) \\ \vdots \\ h(s_k) \end{bmatrix} = \begin{bmatrix} s_0^m & s_0^{m-1} & \dots & s_0^1 & 1 & -s_0^n \cdot h(s_0) & -s_0^{n-1} \cdot h(s_0) & \dots & -s_0^1 \cdot h(s_0) \\ s_1^m & s_1^{m-1} & \dots & s_1^1 & 1 & -s_1^n \cdot h(s_1) & -s_1^{n-1} \cdot h(s_1) & \dots & -s_1^1 \cdot h(s_1) \\ \vdots & & & & & \vdots & & & & \\ s_k^m & s_k^{m-1} & \dots & s_k^1 & 1 & -s_k^n \cdot h(s_k) & -s_k^{n-1} \cdot h(s_k) & \dots & -s_k^1 \cdot h(s_k) \end{bmatrix} \begin{bmatrix} a_m \\ a_{m-1} \\ \vdots \\ a_1 \\ a_0 \\ b_n \\ b_{n-1} \\ \vdots \\ b_1 \end{bmatrix}$$

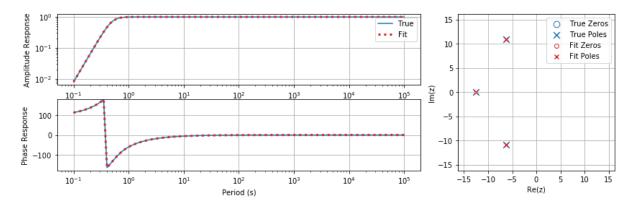
This formulation of a linear least-squares problem is sometimes called the "error-equation" method or the Levy Method in the signal processing literature. However, this form is suboptimal, as the denominator coefficients  $b_n$  are effectively weighted by the response values h(s). A simple method to deal with this bias is to iteratively weight (divide) the linear least squares problem by the denominator term  $d(s_k)$  from the previous iteration until convergence. This is called Sanathanan-Koerner iteration.

```
In [9]: def Fit ZerosPolesGain toFrequencyResponse LLSQ(w, resp, m, n,
                                                         useSKiter=False,
                                                         regularize=False):
            USING Levy Method ("Error-Equation" Method) TO MAKE THIS A LINEAR
            LEAST SQUARES PROBLEM. THEN OPTIONALLY USING Sanathanan-Koerner (SK)
            Iteration TO DEAL WITH THE BIAS INTRODUCED BY THE WAY THE PROBLEM
            IS SET UP. ALSO INCLUDES OPTIONAL REGULARIZATION.
                       : (real) angular frequencies at which we
            w (array)
                              have the system response
            resp (array) : complex system response at angular frequencies w
                      : number of zeros to use in the fit; number of
            m (int)
                              numerator coefficients will be m + 1
            n (int)
                        : number of poles to use in the fit; number of
                              denominator coefficients will be n + 1
            useSKiter (bool) : flips on/off Sanathanan-Koerner (SK) iteration
            regularize (bool) : flips on/off regularization
            11 11 11
            # CONVERT TO COMPLEX (LAPLACE) FREQUENCY, DEFINE USEFUL VALUES
            s = 1.j * w
            num = m + 1
            den = n + 1
            # SET UP ITERATION CONTROLS
            if useSKiter:
                iter max = 100
            else:
                iter max = 1
            tf = None
            iter count = 0
            for it in range(iter max):
                # (1) SET UP DESIGN MATRIX
                # Because of the way this problem is framed, the constant term
                # in the denominator is not included here. See notes above.
                # Basically, the constant term from the denominator is over
                # in the RHS vector.
                g = np.zeros((w.size, num + den - 1), dtype=np.complex128)
                for i in range(num):
                    g[:, i] = s**(num - 1 - i)
                for i in range(den - 1):
                    g[:, i + num] = - resp * s**(den - 1 - i)
                # (2) SET UP WEIGHTS
                # In S-K iteration, the system is weighted by the denominator
                # of theprevious iteration. During the first iteration,
                # set weights = 1.
                if tf is None:
                    weights = np.eye(resp.size)
                else:
```

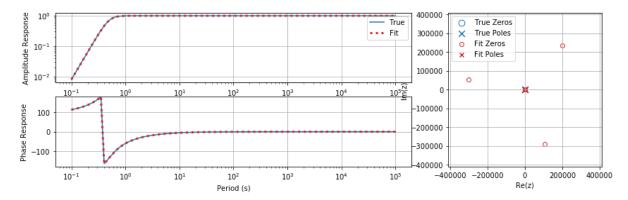
```
weights = np.diag(np.polyval(tf[m+1:], w))
    # (3) COMPUTE THE SVD
    u, sing, vh = np.linalg.svd(np.matmul(weights, g),
                                full matrices=True)
    # (4) REGULARIZE, IF NECESSARY
    if regularize:
        alpha = np.logspace(-4, 4, num=100)
        sing full = np.zeros(u.shape[0])
        sing_full[:sing.size] = sing
        filt = sing_full[np.newaxis, :]**2. / \
               (sing full[np.newaxis, :]**2. +
                alpha[:, np.newaxis]**2.)
        d_bar = np.matmul(u.conj().T, np.matmul(weights, resp))
        err = np.matmul((1. - filt)**2., d_bar**2.) / \
              (resp.size * np.sum(1. - filt, axis=1)**2.)
        alpha_optimal = alpha[np.argmin(err)]
        print("Regularization Parameter: ", alpha_optimal)
        filter_factors = sing**2. / (sing**2. + alpha_optimal**2.)
        print("Max/Min Filter Factors: ", np.amax(filter_factors),
              " / ", np.amin(filter_factors))
    else:
        filter_factors = np.ones(sing.size)
    # (4) CALCULATE THE MODEL PARAMETER
    # Need to concatenate a value of [1] on the end of the model
    # parameter in order to get the full list of transfer function
    # coefficients. The [1] is the constant term in the
    # denominator polynomial.
    model = \
np.matmul(vh.conj().T,
          np.matmul(np.diag(filter factors),
                    np.matmul(np.diag(sing**-1.),
                              np.matmul(u[:, :num+den-1].conj().T,
                                        np.matmul(weights, resp)))))
    tf previous = tf
    tf = np.concatenate((model, np.array([1.])))
    # (5) EVALUATE THE CHANGE FROM THE LAST MODEL PARAMETER
    # Break iteration if the percent change in mean magnitude is <1%
    if tf previous is not None:
        mean value = np.mean(np.absolute(tf previous))
        mean change = np.mean(np.absolute(tf - tf previous))
        if mean change / mean value < 0.01:</pre>
            break
    iter_count += 1
if useSKiter:
    print("Number of S-K iterations: ", iter_count)
# (6) RETURN THE RESULT IN POLES-ZEROS FORM
return signal.TransferFunction(tf[:m+1], tf[m+1:]).to zpk()
```

#### Fitting NIMS B response with linear approach, no S-K iteration and no regularization

#### PARAMETERIZATION: 0 zeros, 3 poles

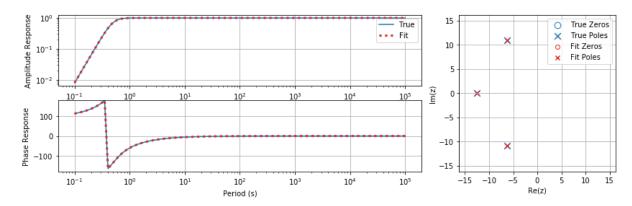


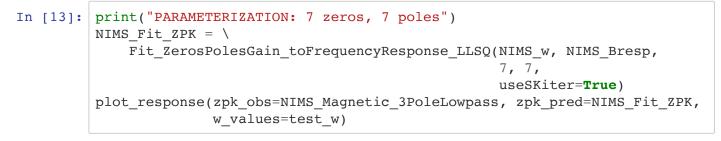
### PARAMETERIZATION: 7 zeros, 7 poles



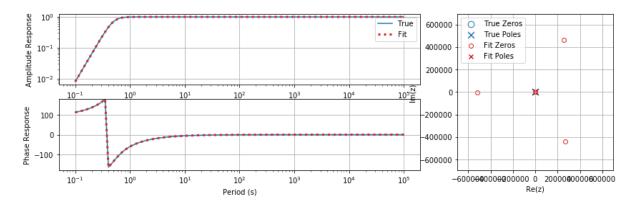
#### Fitting NIMS B response with linear approach, using S-K iteration

PARAMETERIZATION: 0 zeros, 3 poles Number of S-K iterations: 1





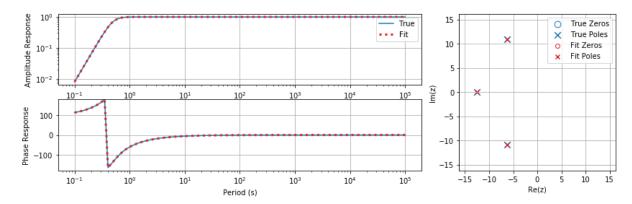
PARAMETERIZATION: 7 zeros, 7 poles Number of S-K iterations: 4



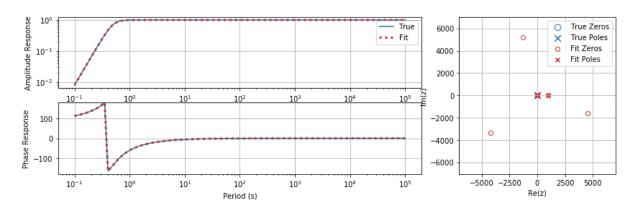
#### Fitting NIMS B response with linear approach, no S-K iteration, regularized

> PARAMETERIZATION: 0 zeros, 3 poles Regularization Parameter: 0.0001

Max/Min Filter Factors: 0.999999999999 / 0.9999999997738326



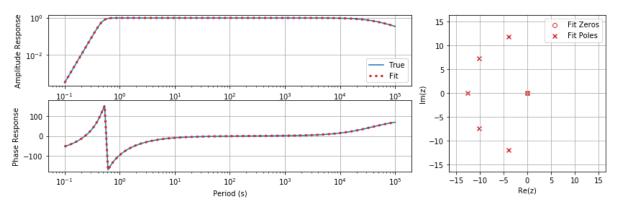
PARAMETERIZATION: 7 zeros, 7 poles
Regularization Parameter: 0.0001
Max/Min Filter Factors: 1.0 / 4.879234765713976e-21



#### Fitting NIMS E response (total) with linear approach, using S-K iteration

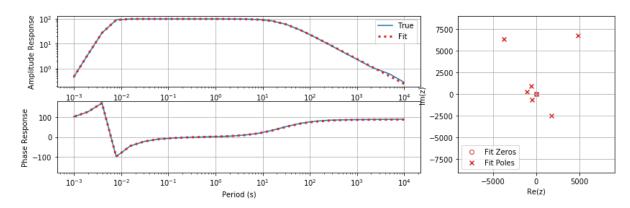
```
print("PARAMETERIZATION: 1 zero, 6 poles")
In [16]:
         test period = np.logspace(-1, 5, num=100)
         test_w = 2. * np.pi / test_period
         NIMS w highpass, NIMS Ehigh = signal.freqresp(NIMS Electric Highpass,
                                                        w=test w)
         NIMS_w_lowpass, NIMS_Elow = signal.freqresp(NIMS_Electric_5PoleLowpass,
                                                      w=test w)
         NIMS w = NIMS w lowpass
         NIMS Eresp = NIMS Ehigh * NIMS Elow
         NIMS EFit ZPK = \
             Fit ZerosPolesGain toFrequencyResponse LLSQ(NIMS w, NIMS Eresp,
                                                          1, 6,
                                                          useSKiter=True)
         plot response(w obs=NIMS w, resp obs=NIMS Eresp, zpk pred=NIMS EFit ZPK,
                       w values=test w)
         print("True Zeros: ", NIMS Electric Highpass.zeros)
         print("Fit Zeros: ", NIMS_EFit_ZPK.zeros)
         print("True Poles: ", NIMS Electric Highpass.poles,
               NIMS Electric 5PoleLowpass.poles)
         print("Fit Poles: ", NIMS_EFit_ZPK.poles)
```

PARAMETERIZATION: 1 zero, 6 poles Number of S-K iterations: 1

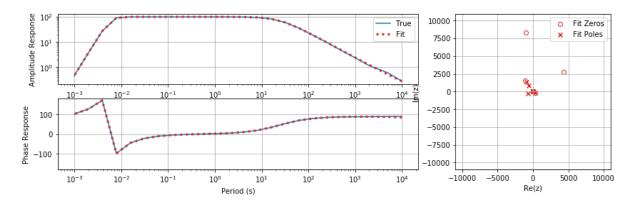


#### Now trying fitting the Zen coil response with the linear approach, no S-K iteration

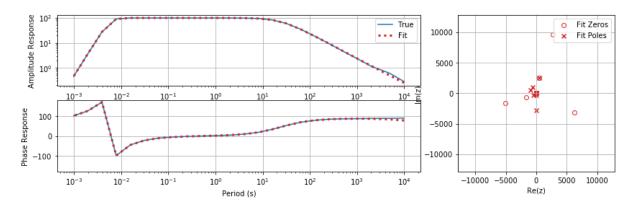
#### PARAMETERIZATION: 1 zero, 7 poles



#### PARAMETERIZATION: 7 zeros, 7 poles

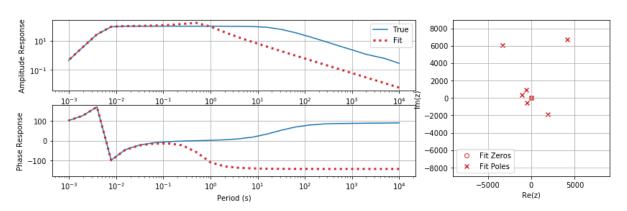


#### PARAMETERIZATION: 10 zeros, 10 poles

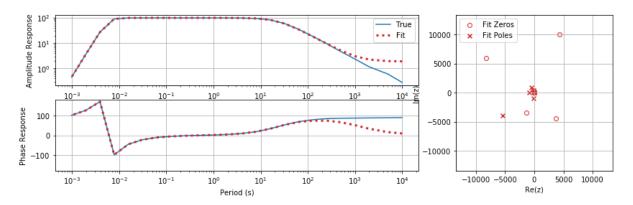


## Now trying fitting the Zen coil response with the linear approach with S-K iteration

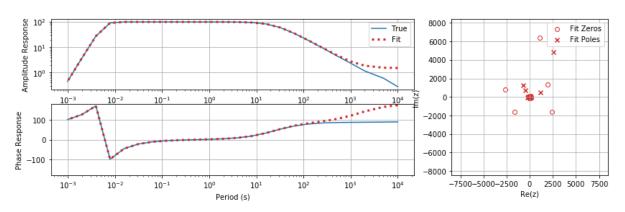
PARAMETERIZATION: 1 zero, 7 poles Number of S-K iterations: 100



PARAMETERIZATION: 7 zeros, 7 poles Number of S-K iterations: 10

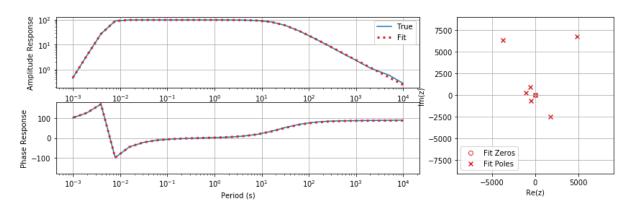


PARAMETERIZATION: 10 zeros, 10 poles Number of S-K iterations: 13

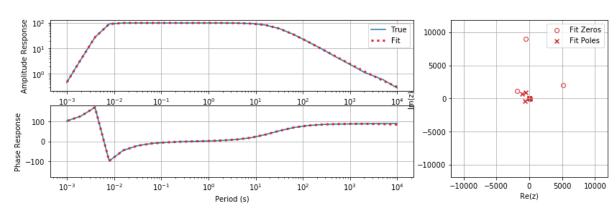


#### Trying to fit Zen coil response with regularized linear least squares, no S-K iteration

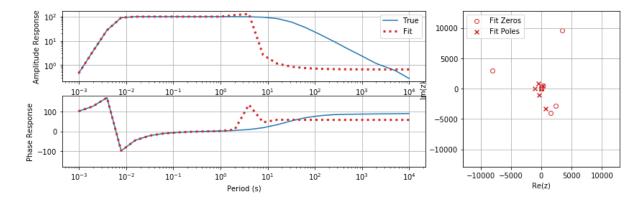
PARAMETERIZATION: 1 zero, 7 poles
Regularization Parameter: 0.0001
Max/Min Filter Factors: 1.0 / 0.999999944819405



PARAMETERIZATION: 7 zeros, 7 poles
Regularization Parameter: 0.007220809018385471
Max/Min Filter Factors: 1.0 / 0.3243396474213219



PARAMETERIZATION: 10 zeros, 10 poles
Regularization Parameter: 14.849682622544666
Max/Min Filter Factors: 1.0 / 1.3403626783077374e-10



Using S-K iteration with the Zen coil response skrews up the fit. The regularization doesn't really seem to help.

And regardless of the use of S-K iteration or regularization, for the Zen coil response, the poles here aren't coming in conjugate pairs, and some of the poles are plotting on the positive side of the real axis. **This might be a problem...?** My impression is that, for stable real systems (no runaway gains, real time series comes in, real time series comes out), the poles/zeros should come in complex conjugate pairs (or fall along the real axis) and should be restricted to the left half plane (-Re). But maybe I'm wrong about that...

# **Vector Fitting**

Vector fitting is another method in the signal processing world to fit a function to complex response data. Instead of using the transfer function (or factored poles/zeros) representation of the system response, vector fitting uses the partial fractions expansion of the rational function:

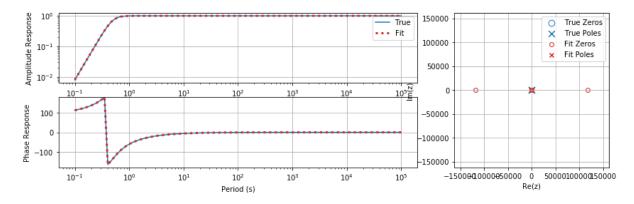
$$h(s) = \sum_{n=1}^{N} \frac{c_n}{s - p_n} + d + s \cdot h$$

where  $c_n$  are the system residues,  $p_n$  are the system poles, d and h are constants, and s is complex frequency.

I'm using someone else's implementation of the algorithm here, so it's sort of a black box, and it also is perhaps overkill for our relatively simple systems. (VF seems to be used on complicated functions, not the simple functions that we deal with: flat in some pass band and then falling off with frequency on either side.)

#### **Vector Fitting with NIMS B Response**

```
poles: -12.5666 + 0j, -11.5852 + 0j, -6.34538 + -10.8346j, -6.34538 + 1 0.8346j, -6.28322 + -10.8827j, -6.28322 + 10.8827j residues: 12.5644 + 0j, 0.00181701 + 0j, 0.00128419 + 0.0151252j, 0.001 28419 + -0.0151252j, -6.28442 + 3.61257j, -6.28442 + -3.61257j offset: -1.31964e-08 slope: 1.03931e-10
```



```
# poles: 6; # zeros: 5
```

#### **Vector Fitting with Zen coil response**

# poles:

4 ;

# zeros:

```
poles, residues, d, h = \
In [27]:
                 vectfit.vectfit_auto(zen_complex_response,
                                            1.j * zen angular frequencies,
                                            n poles=2)
            num , den = signal.invres(residues, poles, [h, d])
            Zen_VF_ZPK = signal.TransferFunction(num_, den_).to_zpk()
            plot_response(w_obs=zen_angular_frequencies,
                              resp_obs=zen_complex_response,
                              zpk_pred=Zen_VF_ZPK, w_values=zen_response[:, 0])
            print("# poles: ", Zen_VF_ZPK.poles.size, "; # zeros: ",
                    Zen VF ZPK.zeros.size)
            poles: -1072.29 + 0j, -538.767 + -919.537j, -538.767 + 919.537j, -0.261
            917 + 0j
            residues: 111274 + 0j, -55091.6 + 29648.9j, -55091.6 + -29648.9j, -26.0
            316 + 0j
            offset: -0.107995
            slope: 3.11315e-05
             Amplitude Response
                                                                            8000
                                                                      True
                                                                                                0
                                                                                                  Fit Zeros
                                                                   • • • Fit
                                                                                                  Fit Poles
                                                                            6000
               10¹
                                                                            4000
               10°
                                                                            2000
                   10^{-3}
                          10^{-2}
                                  10^{-1}
                                                10<sup>1</sup>
                                                        10<sup>2</sup>
                                                               10<sup>3</sup>
                                                                       10^{4}
                                                                           -2000
            Phase Response
               100
                                                                            -4000
                0
                                                                            -6000
              -100
                                                                            -8000
                   10-3
                                                                                    -5000
                                                                                                 5000
                           10-2
                                  10-1
                                                               10<sup>3</sup>
                                                                      104
                                         100
                                                101
                                                        10<sup>2</sup>
                                           Period (s)
```

Although I'm telling the VF algorithm to start with just a few poles, it is automatically adding more poles based on some internal criteria. This is resulting in overfitting the NIMS B response curve with too many poles (and zeros). For the fit to the Zen coil response, the poles are at least coming out in conjugate pairs, but the algorithm is also producing zeros in the right half of the complex plane.

However, this is again at least producing a poles-zeros representation of the response function that captures the amplitude and phase behavior correctly...

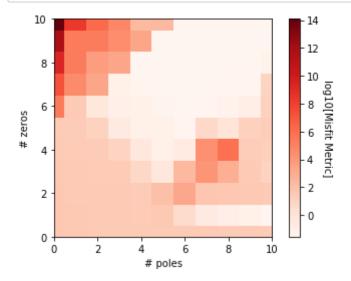
# **Automated Poles/Zeros Fitting**

This is a simple code snippet that could be used to pick out a parameterization for a poles-zeros-gain representation of a complex response lookup table. Basic idea in my implementation here:

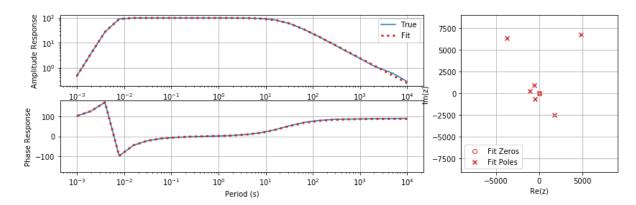
- 1. Loop over a number of poles and zeros (up to an arbitrary total number).
- 2. Fit the given number of poles/zeros to the lookup table complex values.
- 3. Evalute misfit of the fit via some metric.
- 4. Once the misfits from the all poles/zeros combinations have been evaluated, pick the fit that has the least number of zeros while also reaching some target misfit level.

Picking the parameterization with the least number of zeros makes the most sense to me. With the types of system responses we generally deal with (flat in some pass band, then steeply rolling of on the sides), we're most likely going to have more poles than zeros (e.g., Buttersworth 3 pole, 5 pole, etc.). Here, I'm just using the simplest implementation of the linear least squares problem (no iterative reweighting, no regularization).

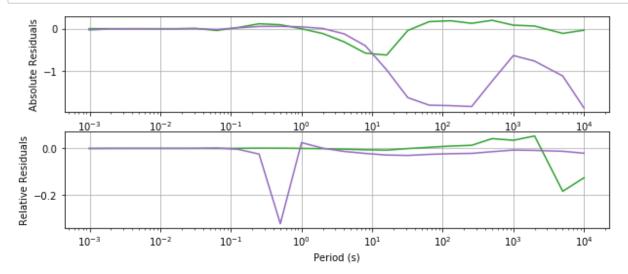
/Users/bmurphy/.conda/envs/geo/lib/python3.7/site-packages/scipy/signa l/filter\_design.py:1619: BadCoefficients: Badly conditioned filter coef ficients (numerator): the results may be meaningless "results may be meaningless", BadCoefficients)



#### CHOSEN PARAMETERIZATION TO REACH DESIRED MISFIT: 1 zeros, 7 poles



```
In [33]:
         w pred, resp pred = signal.freqresp(Zen_ZPK, w=zen_angular_frequencies)
         amplitude residuals = \
             np.absolute(resp_pred) - np.absolute(zen_complex_response)
         phase_residuals = \
             np.angle(resp pred, deg=True) - np.angle(zen_complex_response,
                                                       deg=True)
         fig = plt.figure(figsize=(10, 4))
         ax abs = fig.add subplot(211)
         ax_abs.plot(zen_period, amplitude_residuals, c='tab:green')
         ax_abs.plot(zen_period, phase_residuals, c='tab:purple')
         ax_abs.set_xscale('log')
         ax_abs.set_ylabel('Absolute Residuals')
         ax_abs.grid()
         ax_rel = fig.add_subplot(212)
         ax_rel.plot(zen_period,
                      amplitude residuals/np.absolute(zen_complex_response),
                      c='tab:green')
         ax_rel.plot(zen_period,
                      phase_residuals/np.angle(zen_complex_response, deg=True),
                      c='tab:purple')
         ax_rel.set_xscale('log')
         ax_rel.set_xlabel('Period (s)')
         ax_rel.set_ylabel('Relative Residuals')
         ax rel.grid()
         plt.show()
```



# **Concluding Thoughts/Questions**

So, using either linear least squares or vector fitting, we can derive a poles-zeros-gain representation of the Zen coil response lookup table. The question is, does it matter if the poles/zeros aren't coming out in conjugate pairs and if any poles/zeros appear in the right half (+Re) of the complex plane. My (perhaps erroneous) impression is that, for the real systems we deal with (real-valued time series comes in, real-valued time series comes out, no instabilities or runaway gains in the system), poles/zeros should come in complex conjugate pairs (or fall along the real axis) and should be restricted to the left half plane (-Re). But maybe I'm wrong about that.

One way or another, we have a poles/zeros/gain representation at some level of misfit to the lookup table response, and maybe (hopefully) that's good enough for archiving with IRIS (even if the fit is overparameterized with too many poles/zeros, or even if the fit isn't 100% perfect).

### References

https://cdn.intechopen.com/pdfs/57667.pdf (https://cdn.intechopen.com/pdfs/57667.pdf)

#### **Linear LS Approach**

https://www.dsprelated.com/freebooks/filters/Filter Design Minimizing L2.html (https://www.dsprelated.com/freebooks/filters/Filter Design Minimizing L2.html)

https://www.dsprelated.com/showcode/20.php (https://www.dsprelated.com/showcode/20.php)

### **Vector Fitting**

https://arxiv.org/pdf/1908.08977.pdf (https://arxiv.org/pdf/1908.08977.pdf)

https://github.com/PhilReinhold/vectfit\_python (https://github.com/PhilReinhold/vectfit\_python)