Yes! From analyzing the image, I can extract the key information regarding the **transfer function** transformation from the Laplace domain (H(s)) to the Z-domain (H(Z)) for a Digital PLL (DPLL).

## **Key Points from the Image:**

- 1. Laplace-Domain Representation:
  - The continuous-time transfer function is given as:

$$H(s) = rac{2\eta \omega_n s + \omega_n^2}{s^2 + 2\eta \omega_n s + \omega_n^2}$$

- This corresponds to an analog second-order system.
- 2. Discretization Using the Z-Transform:
  - The discrete-time **Z-domain transfer function** is:

$$H(Z) = rac{C_2(Z-1) + C_1}{(Z-1)^2 + C_2(Z-1) + C_1}$$

- ullet Here, Z replaces s as the complex frequency variable.
- 3. Mapping from Continuous to Discrete:
  - The transformation from H(s) to H(Z) is done assuming **high sampling frequency** ( $f_s\gg f_n$ ), which allows the approximations:

$$C_2=2\eta\omega_n T, \quad C_1=rac{C_2^2}{4\eta^2}$$

- Where:
  - $T=rac{1}{f_s}$  (sampling period),
  - $\omega_n=2\pi f_n$  (natural frequency),
  - $\eta$  is the damping factor.
- 4. Relation to the Unit Circle:
  - In discrete-time systems, the Z-transform is expressed as:

$$Z=e^{j\Omega}$$

- This means that the digital filter operates in the unit circle of the complex plane, where:
  - $\Omega$  is the discrete-time angular frequency,
  - The poles and zeros define system stability and frequency response.

• The poles and zeros define system stability and frequency response.

## Interpretation:

- ullet The analog transfer function (H(s)) describes a continuous second-order PLL.
- The discrete transfer function (H(Z)) is obtained using a Z-transform, mapping the system into a digital filter.
- ullet The substitution  $Z=e^{j\Omega}$  places the analysis on the **unit circle**, where digital frequency responses are evaluated

Would you like me to generate the frequency response of H(Z) using the given parameters?  ${\mathscr J}$ 



I have no problem with the transformation but I am no sure for the meaning of the variable Omega n relative to T. May be at this point you can have some insight that are missing in the paper?

Good question! Let's clarify the meaning of  $\omega_n$  (natural frequency) relative to the sampling period T.

# Understanding $\omega_n$ in Continuous and Discrete Time

- 1. Continuous-Time Interpretation (s-Domain)
  - The **natural frequency** of the system is given by:

$$\omega_n = 2\pi f_n$$

- Where:
  - $f_n$  is the natural frequency in Hz.
  - $\omega_n$  is the natural angular frequency in rad/s.
  - This applies to the analog domain.
- 2. Discrete-Time Interpretation (Z-Domain)

• The system is sampled with a period:

$$T=rac{1}{f_s}$$

• The sampling frequency is related to the discrete-time variable  $\Omega$ , which represents **digital** angular frequency in the unit circle:

$$\Omega = \omega_n T$$

- In digital systems:
  - $\Omega$  is in radians per sample.
  - It is the discrete-time equivalent of  $\omega_n$ .

#### **Key Relationship:**

- In continuous time,  $\omega_n$  is given in radians per second.
- In discrete time, the equivalent frequency is  $\Omega=\omega_n T$  , which is in radians per sample.
- Since T is the sampling interval, the discrete-time system operates in a normalized frequency space.

# Possible Missing Insights in the Paper

- 1. Interpretation of  $\omega_n$  in Discrete-Time Analysis
  - ullet The paper assumes  $T\ll 1$ , meaning the system is sampled at a high rate.
  - However, it does not explicitly mention the transformation:

$$\Omega = \omega_n T$$

- This is important because it scales the frequency response in the digital domain.
- 2. Normalized Frequency in Digital Systems
  - The paper defines  $\omega_n=2\pi f_{n_i}$  but it does not explicitly relate it to the **Nyquist frequency**.
  - In digital filtering, frequencies are often normalized:

$$\hat{\Omega} = rac{\omega_n}{f_s} = rac{\Omega}{2\pi}$$

• This normalization ensures that digital frequencies are bounded between:

$$-\pi \leq \Omega \leq \pi$$

• This might be missing in the paper's explanation.

## **Conclusion**

- The missing step is the transition from continuous-time frequency ( $\omega_n$ ) to discrete-time frequency ( $\Omega=\omega_n T$ ).
- If the system response is analyzed in digital form, we must consider the effect of **sampling** and interpret  $\Omega$  on the **unit circle**.

Would you like me to illustrate this relationship with a graph? III



